

Lecturer 6, 7, 8, 9

# Radio Communication Systems

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# Outline

2

- Introduction
- Types of Digital Modulation
  - Frequency Shift Keying FSK
    - MSK Minimum Shift Keying
  - Amplitude Shift Keying ASK
  - Phase Shift Keying PSK
- M-ary PSK Encoding
  - Quadrature QPSK
  - 8-PSK
- QAM: (8-QAM)
- Carrier Recovery Circuits

# Digital Radio

## ➤ Why Digital ?

- Ease of processing,
- Ease of multiplexing, and
- Noise immunity.

## ➤ All Digital Communications

- Transmission, reception and processing of information.

## ➤ Increasing of Information Capacity

- No of independent symbols that can be carried through system in a given time.

# Information Capacity

- **1928 Hartley introduces useful rule:**
  - Capacity  $C$  is proportional to both the bandwidth  $B$  and the time  $T$ :

$$C \sim B T$$

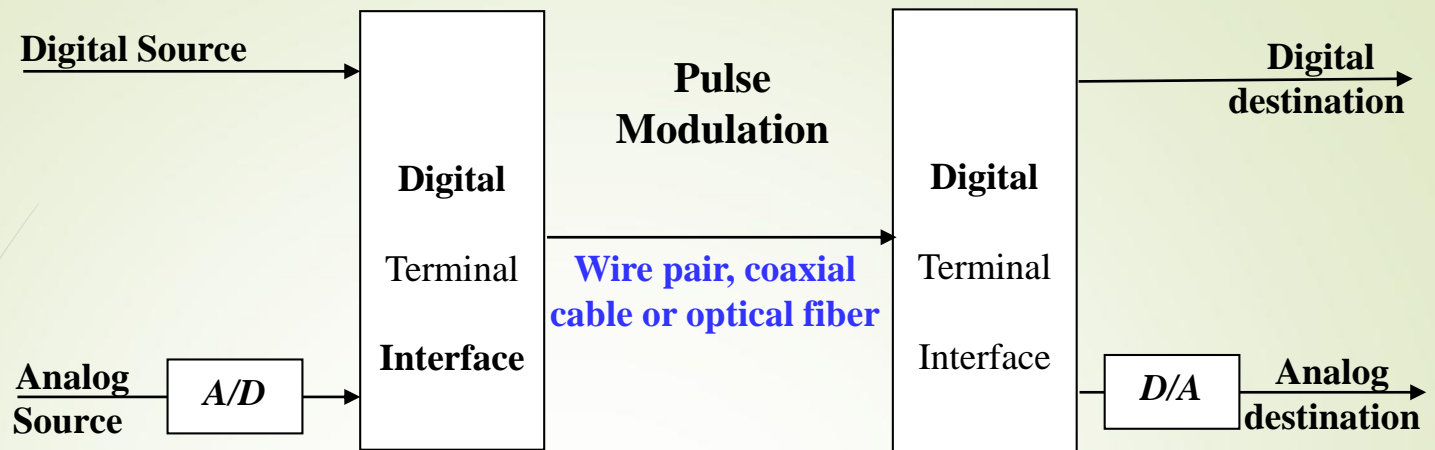
- **1948 Shannon published a limit for  $C$ :**

$$C \leq B \log_2 \left( 1 + \frac{S}{N} \right) = 3.32 B \log_{10} \left( 1 + \frac{S}{N} \right)$$

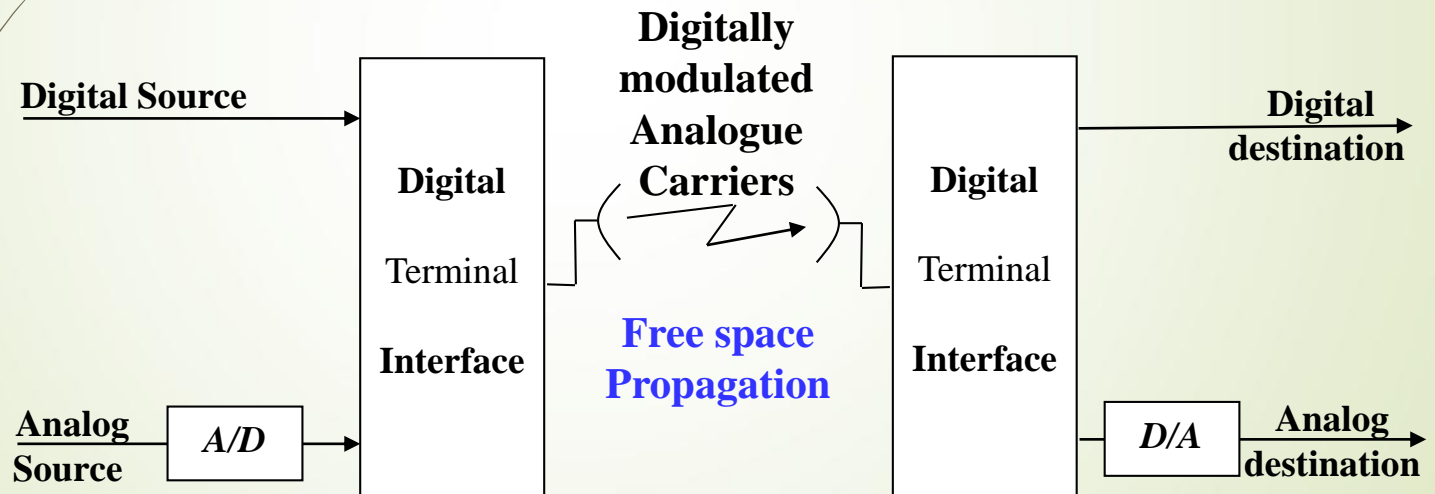
- $S/N = 1000$  (30 dB),  $B = 2.7\text{kHz}$ ,  $C$ :  
 $C \leq 2700 \log_2(1 + 1000) \leq 26.9 \text{ kbps}$

# Limit Misunderstood

- Above example may be true, but it cannot be done with a binary system.
- To achieve 26.9 kbps through 2.7 kHz channel, each symbol must contain more than one bit of information.
- So, to achieve Shannon limit, digital transmission systems that have more than two output conditions (symbols) must be used.



**(a) Baseband Transmission**



**(b) Digital Radio Transmission**

# Types of Modulation

■ تعديل إزاحة السعة ASK Amplitude Shift Keying

■ تعديل إزاحة التردد FSK Frequency Shift Keying

■ تعديل الإزاحة الدنيا MSK Minimum Shift Keying

■ تعديل الإزاحة الدنيا الجاوسي GMSK Gaussian Minimum Shift Keying

■ تعديل إزاحة الطور PSK Phase Shift Keying

■ تعديل إزاحة الطور الثنائي BPSK Binary Phase Shift Keying

■ تعديل إزاحة الطور التفاضلي DPSK Differential Phase Shift Keying

■ تعديل إزاحة الطور متعدد المستويات M-ary Phase Shift Keying

■ تعديل إزاحة الطور التعامدي QPSK Quadrature Phase Shift Keying

■ تعديل إزاحة الطور الثماني 8PSK Eight Levels Phase Shift Keying

■ تعديل السعة التعامدي QAM Quadrature Amplitude Modulation

**FSK**

**Frequency Shift  
Keying**



# FSK Transmitter Signal

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Simple, low performance.

Constant envelope angle modulation.

$$v_{FSK}(t) = V_c \cos \left[ \left( \omega_c + f_m(t) \frac{\Delta\omega}{2} \right) t \right]$$

$f_m(t)$  binary digital modulating signal

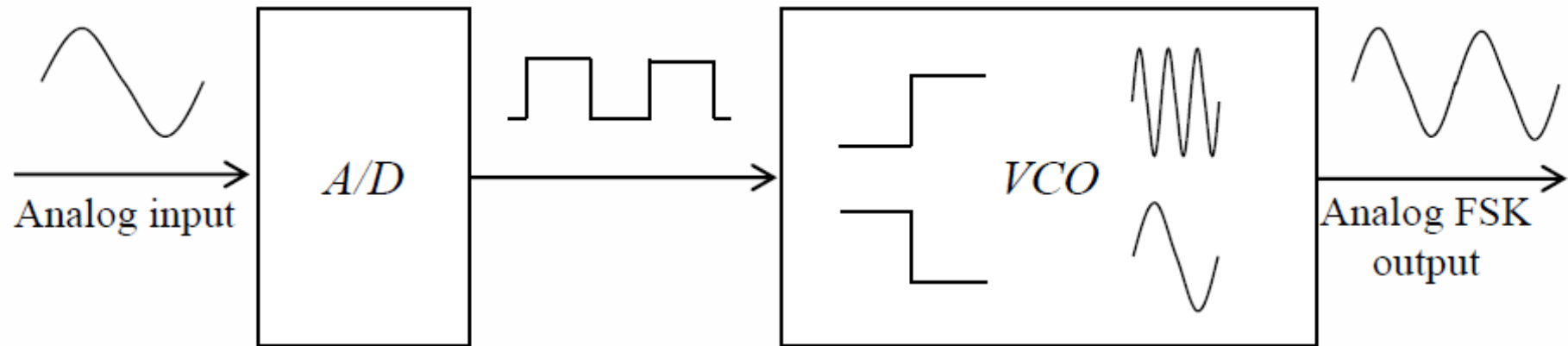
$V_c, \omega_c$ , carrier amplitude, frequency

Carrier frequency shifts between  $\omega_c \pm \Delta\omega/2$

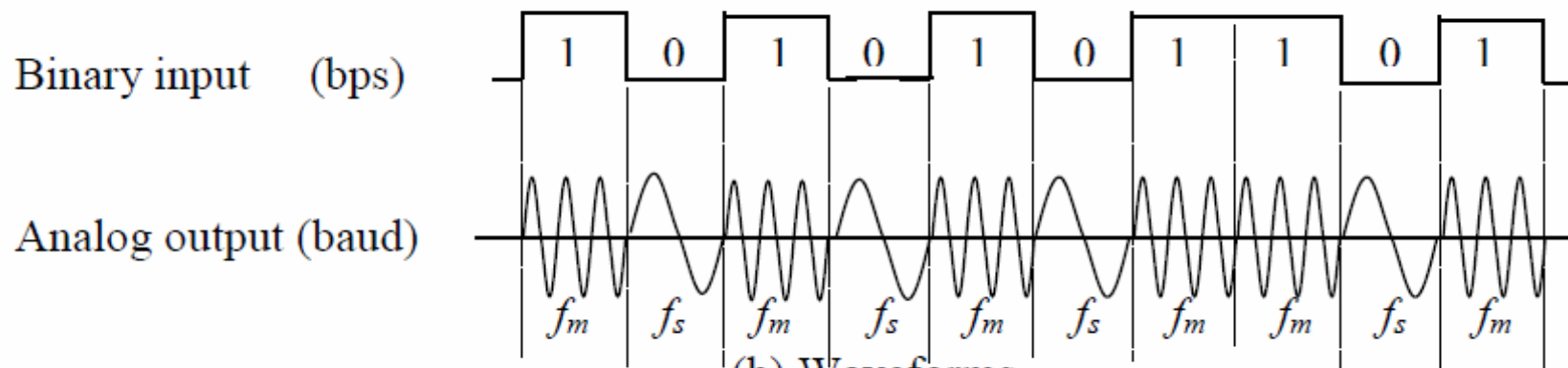
Shift rate equals the input bit rate  $f_b$  b/s.

# FSK Transmitter

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(a) FSK Transmitter



(b) Waveforms

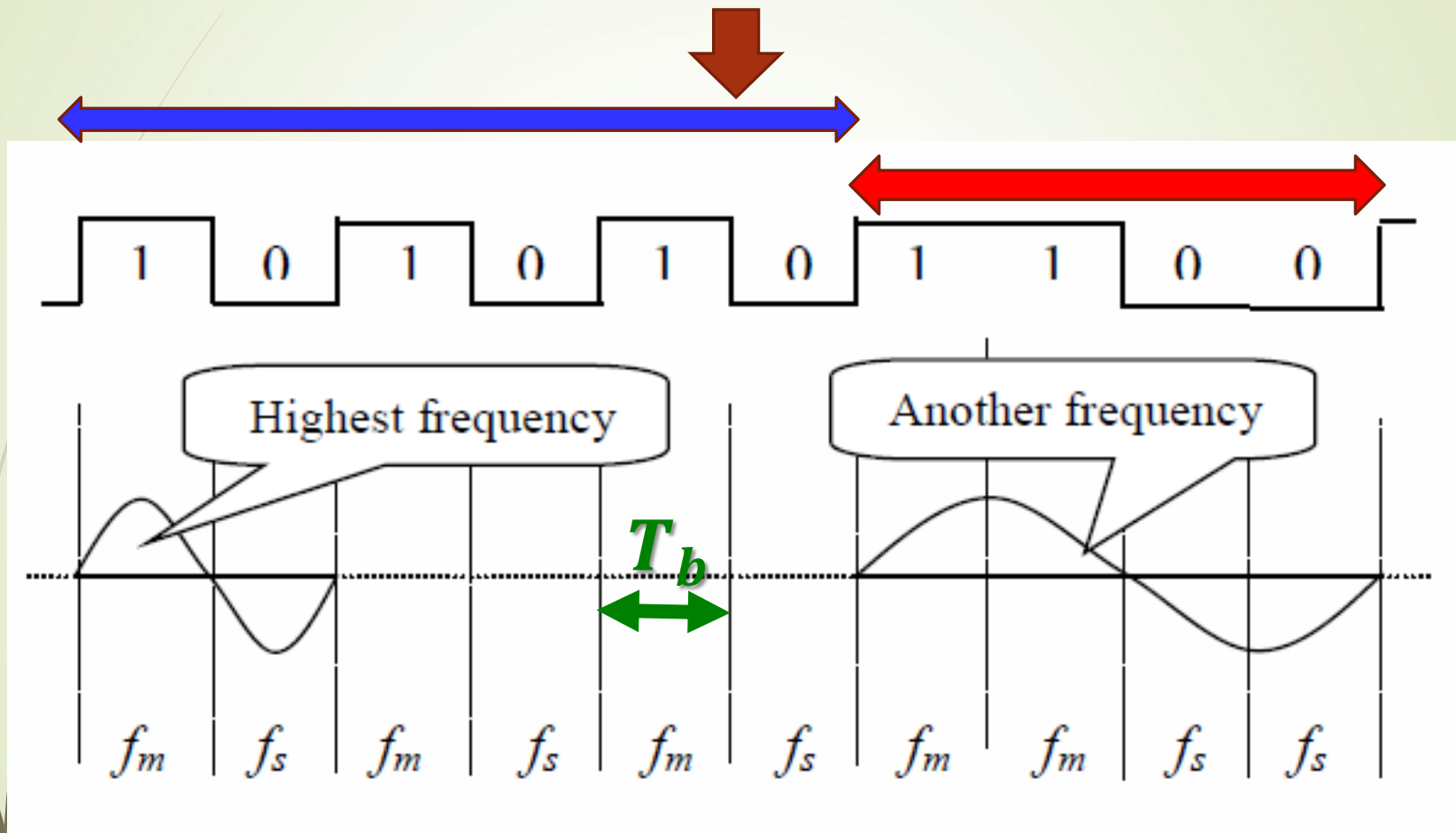
Fig 2.2 Binary FSK Modulator

# Bit and Baud Rates

- **Bit rate**, in bits per second,
  - Is the rate of change at the input to the modulator.
- **Baud rate**, in symbol per sec
  - Is the rate of change at the output of the modulator and
  - Is equal to the reciprocal of the time of one output signaling element (termed as symbol).
- So, baud is the line speed in symbols per second.

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# Possible frequencies



# Highest Modulating Frequency

- If bit width is  $T_b$ , bit rate will be  $f_b = \frac{1}{T_b}$
- Fastest rate occurs when input is a series of alternating 1's and 0's:
- If fundamental frequency is considered, highest modulating frequency is one-half the input bit rate.

$$f_m = \frac{f_b}{2}$$

# Modulation Index of FSK

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- Peak frequency deviation  $\Delta f$  is one half the difference between  $f_m$  and  $f_s$ :

$$\Delta f = \frac{f_m - f_s}{2}$$

- Formula for modulation index used in FM is also valid for binary FSK as:

$$MI = \frac{\Delta f}{f_m} = \frac{\frac{f_m - f_s}{2}}{\frac{f_b}{2}} = \frac{f_m - f_s}{f_b}$$

- MI is kept below 1.0 for narrow band FM.
- BW is determined from Bessel functions table.
- MI 0.5 and 1.0, either two or three sets of significant side frequencies are generated.
- Thus, minimum BW is two or three times the bit rate.

# Bandwidth of Binary FSK

- BW for FSK signal is given by Carson's rule in terms of the frequency deviation and the bandwidth of the digital modulation

$$BW_{FSK} = 2(\Delta f + B)$$

- For alternating 1 and 0, the bandwidth equals the bit rate  $B = R$ :

$$BW_{FSK} = 2(\Delta f + R)$$

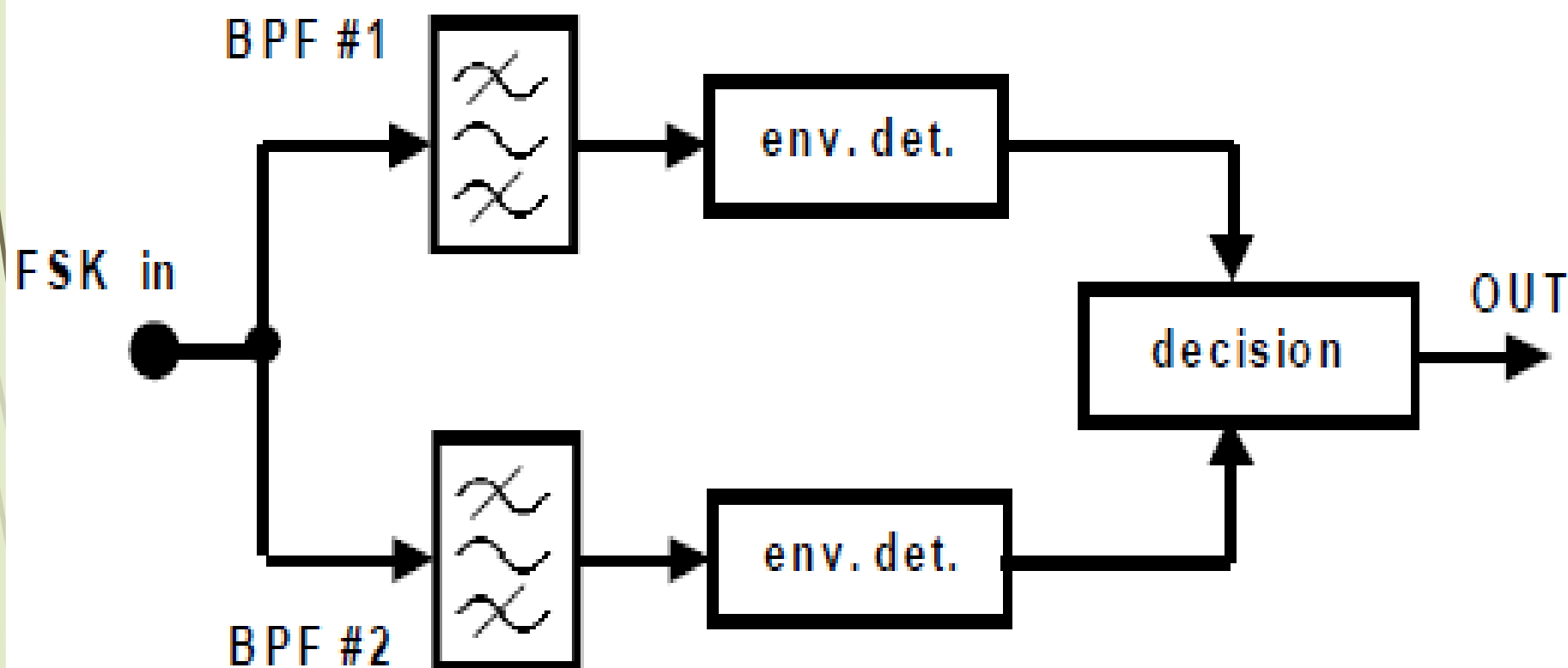
# Receiver Binary FSK

- **Noncoherent Detection:**
  - We do not have knowledge of the carrier.
  - Signal coming is divided into two BPF and envelope detectors.
  - Finally, binary restoration circuit.
- **Coherent detection:**
  - We need a complete knowledge of the exact carrier frequency on reception.
  - Received signal is applied into two multipliers, at  $f_1$  and  $f_0$ , then to LPF.
  - Finally binary restoration circuit.



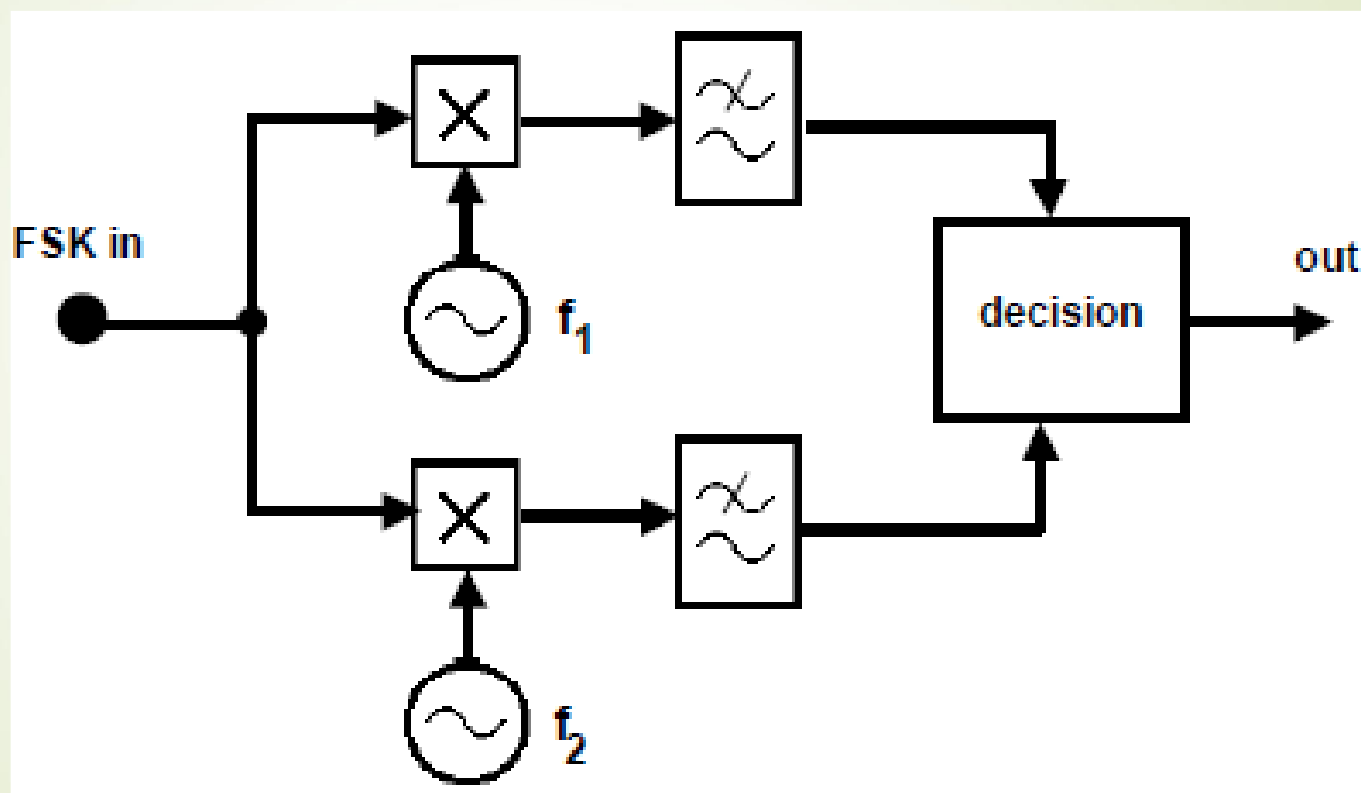
## FSK

## Noncoherent Detector



## FSK

## Synchronous Detector



# Applications of FSK

- Binary FSK has a poorer error performance than PSK or QAM.
- Its use is restricted to low-performance, low-cost, asynchronous data **modems** that are used for **data communication** over analogue, voice band **telephone lines**.

# Bell 103-type FSK Modem

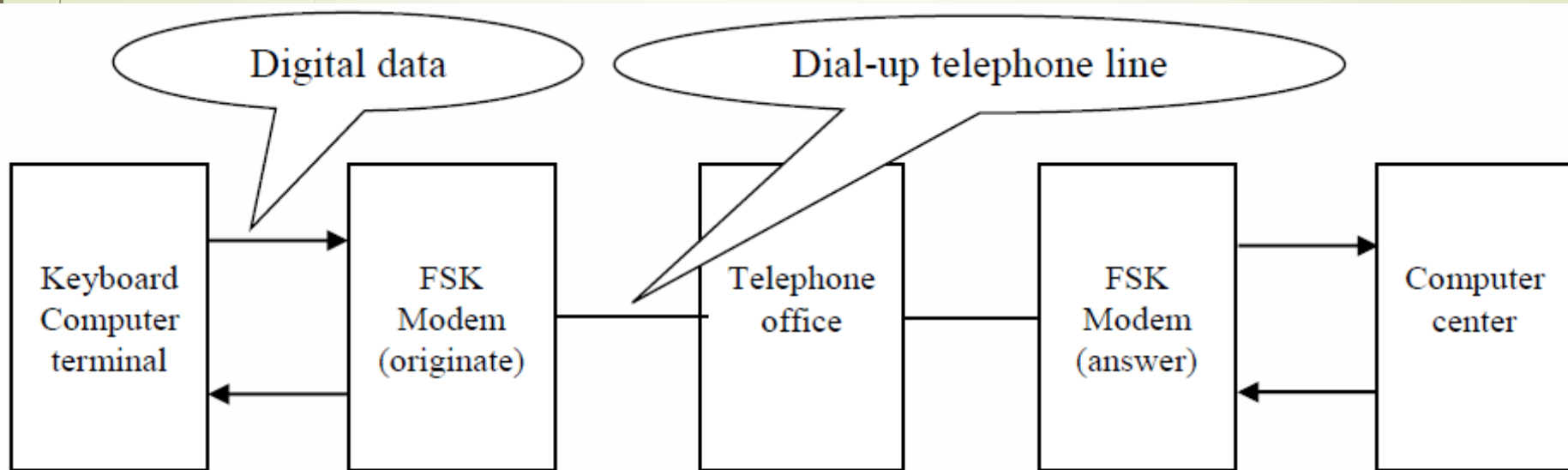


Table 2.1 Mark and Space Frequencies for the Bell Type 103 Modem

	Data	Originate Modem	Answer Modem
Transmit frequencies	Mark (binary 1)	$f_1 = 1270 \text{ Hz}$	$f_1 = 2225 \text{ Hz}$
	Space (binary 0)	$f_2 = 1070 \text{ Hz}$	$f_2 = 2025 \text{ Hz}$
Receive frequencies	Mark (binary 1)	$f_1 = 2225 \text{ Hz}$	$f_1 = 1270 \text{ Hz}$
	Space (binary 0)	$f_2 = 2025 \text{ Hz}$	$f_2 = 1070 \text{ Hz}$

## Bell 103-type FSK Modem

- Keyboard-type computer terminals are often used for communication with a remote computer via dial-up telephone lines.
- Dial-up means that the computer terminal user calls the computer facility on a telephone and uses the telephone connection for data communication.
- Modem (modulator and demodulator) is connected to the phone line at each end as shown
- Two FSK frequency bands are used (one around 1 kHz and another around 2 kHz) so that it is possible to talk and listen simultaneously (full-duplex).
- The standard mark and space frequencies are shown in Table where the peak to peak deviation is  $2\Delta F = 200 \text{ Hz}$

# **MSK**

# **Minimum Shift Keying**

# Minimum Shift Keying

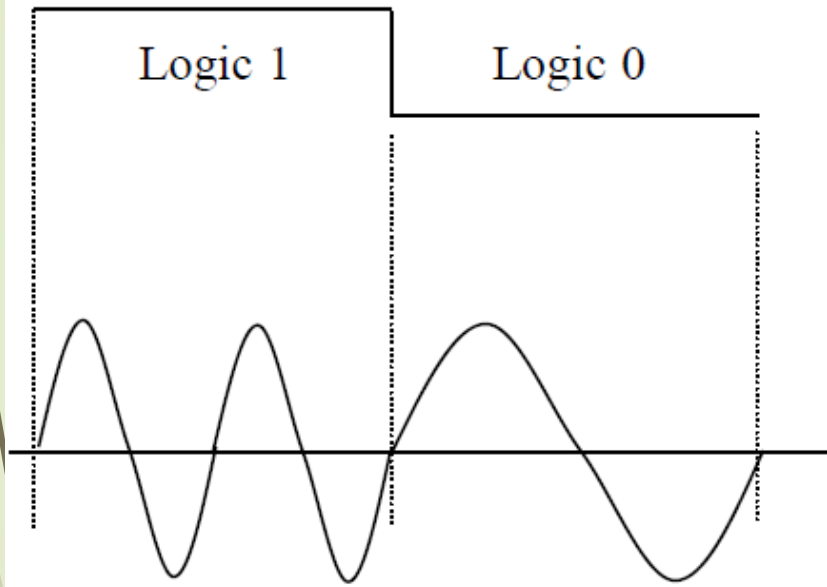
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- ❑ MSK is a continuous phase FSK keying (CPFSK).
- ❑ MSK is FSK except mark and space frequencies are synchronized with input binary rate.
- ❑ Synchronous means precise time relationship.
- ❑ Mark and space frequencies are separated from center frequency by odd multiple of one-half  $f_b$

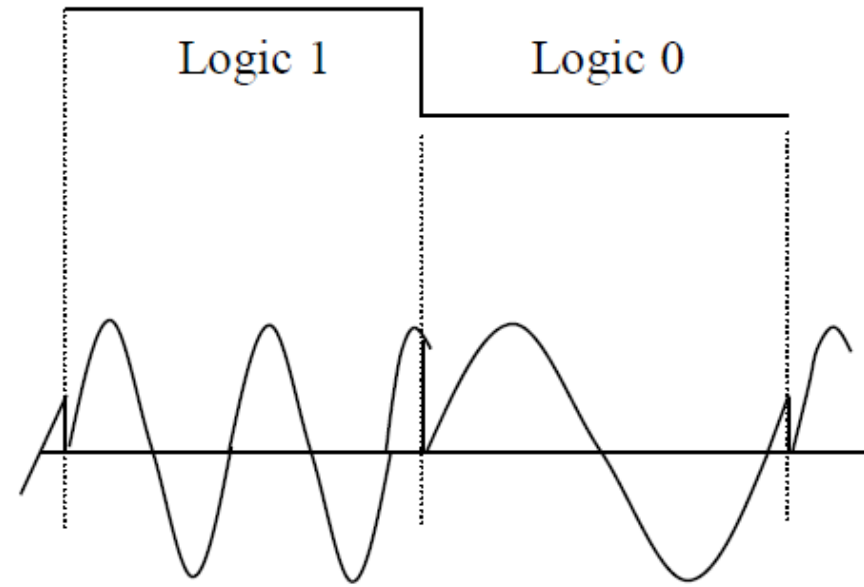
$$f_m \text{ and } f_s = n \frac{f_b}{2}$$

- ❑ MSK has a better bit error performance than FSK for a given signal to noise ratio.
- ❑ MSK has less bandwidth than FSK
- ❑ However, it requires synchronizing circuits and is therefore more expensive to implement.

# MSK versus FSK



(a) Continuous Phase MSK



(b) Non-continuous FSK

Fig.2.8 Comparison of the Phase Continuity between MSK and FSK



# ASK

# Amplitude Shift Keying

# Amplitude Shift Keying

- ❑ In ASK, amplitude of carrier switches between; zero (Off state) and some amplitude (On state)
- ❑ Such systems are termed on-off-keyed systems OOK.
- ❑ Spectrum of OOK depend on the particular binary sequence to be transmitted. However, the amplitude modulated OOK is the DSB.SC given by:

$$f_{OOK}(t) = f_{ASK}(t) = A f(t) \cos \omega_c t$$

- ❑ Spectrum of OOK signal is given as:

$$F_{OOK}(\omega) = F_{ASK}(\omega) = \frac{A}{2} [F(\omega - \omega_c) + F(\omega + \omega_c)]$$

# OOK Waveforms

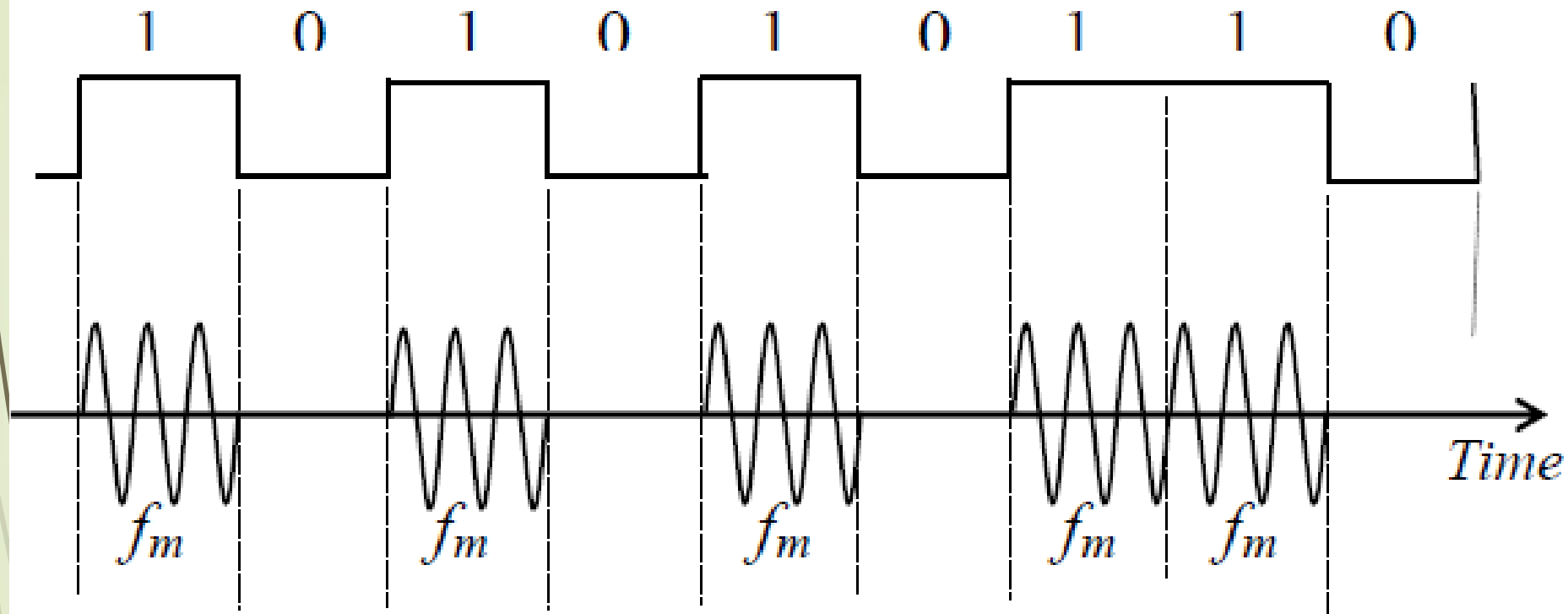


Fig.2.9 ASK or OOK Signal

# Spectrum of OOK

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- ❑ Assume the digital signal  $f(t)$  is rectangular pulse (special case of binary in which all symbols are 0 except for one 1).
- ❑ For a pulse of amplitude  $A$  and duration  $T$ , the spectrum of OOK modulator is given by:

$$F_{OOK}(\omega) = \frac{AT}{2} \left[ \frac{\sin(\omega - \omega_c)T/2}{(\omega - \omega_c)T/2} + \frac{\sin(\omega + \omega_c)T/2}{(\omega + \omega_c)T/2} \right] = \frac{AT}{2} \left[ \text{Sa} \left\{ \frac{(\omega - \omega_c)T}{2} \right\} + \text{Sa} \left\{ \frac{(\omega + \omega_c)T}{2} \right\} \right]$$

- ❑ For alternating 1's and 0's, spectrum is  $(\sin x) / x$ .
- ❑ So, spectrum of pulse of width  $T$  and period  $2T$  which is translated to frequency  $f_c$  as in Fig.2.10

# Spectrum of Periodic OOK

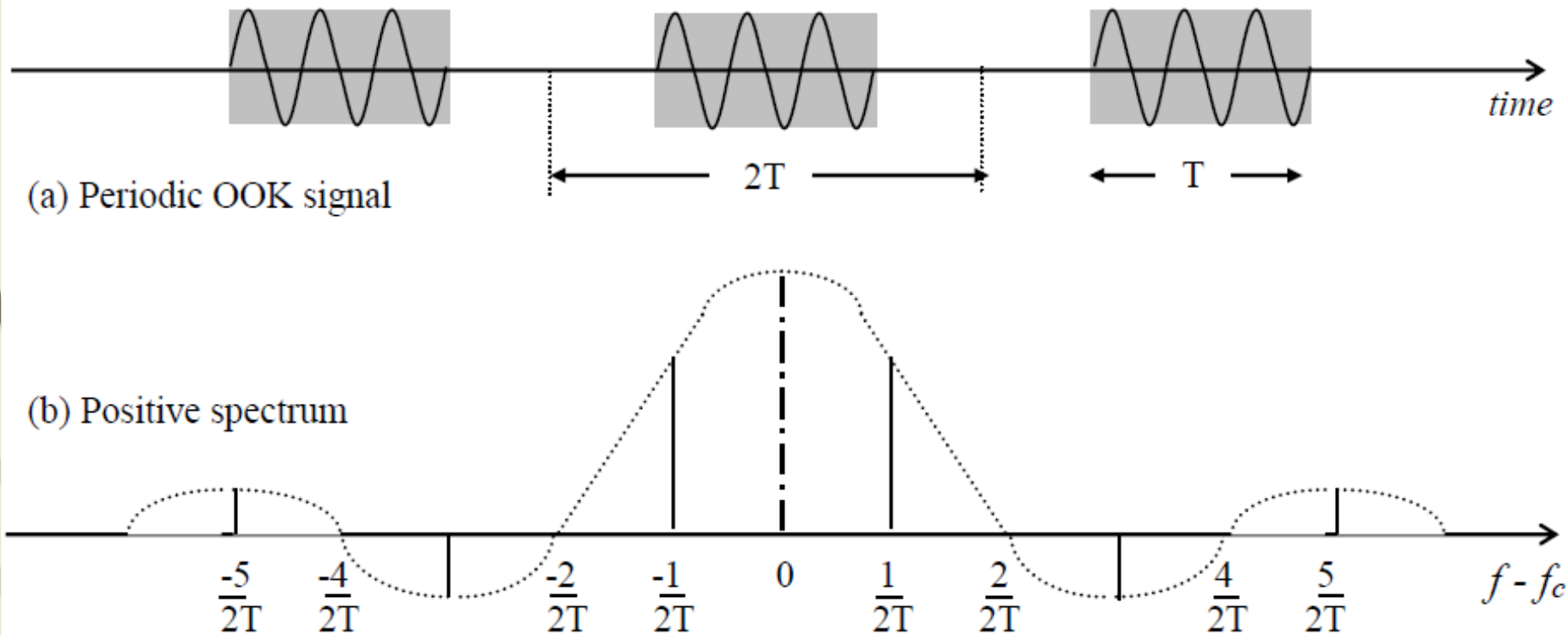


Fig.2.10 Spectrum of Periodic OOK Signal

# **PSK**

# **Phase Shift**

# **Keying**

# Phase Shift Keying

- ❑ PSK is similar to phase modulation PM except that its input gives rise to a limited number of output phases.
- ❑ With binary BPSK two output phases are possible for a single carrier frequency. One phase represents a logic 1 and the other represents a logic 0.
- ❑ With carrier amplitude  $V_c$  and frequency  $\omega_c$  PSK voltage for binary digital modulating signal  $f(t)$  is:

$$v(t) = V_c f(t) \sin \omega_c t = \begin{cases} +V_c \sin \omega_c t & \text{if } f(t) = +1 \\ -V_c \sin \omega_c t & \text{if } f(t) = -1 \end{cases}$$

- ❑ So, the carrier amplitude remains constant, whereas its phase shifts by  $180^\circ$ .
- ❑ Recall, carrier phase shift rate equals input bit rate.

# BPSK Waveforms

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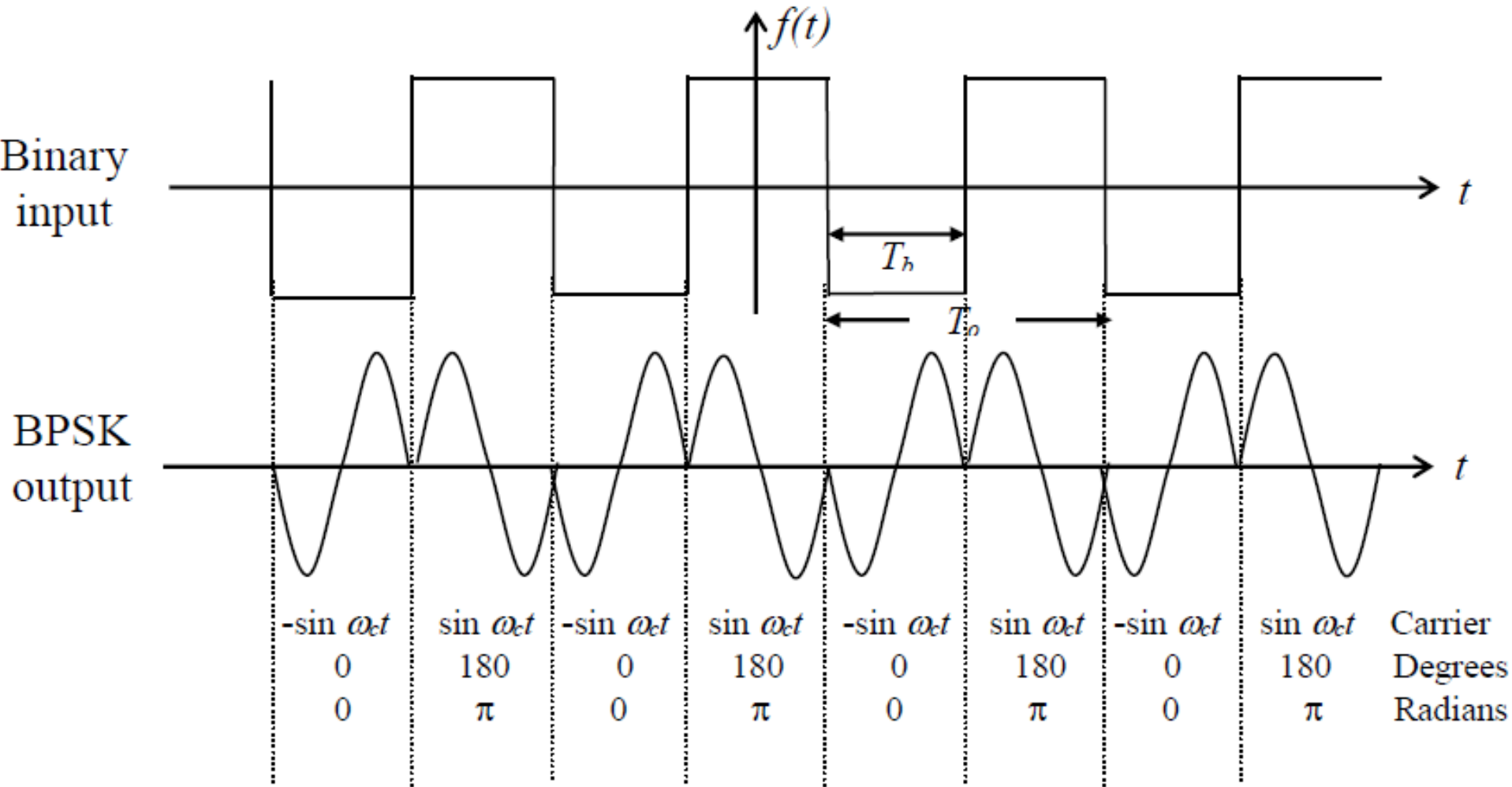


Fig.2.14 Output of BPSK Modulator



# PSK Modulator

- ❑ Simplified block diagram of BPSK is shown in Fig.2.11
- ❑ Balanced modulator acts as a phase reversing switch.
- ❑ Carrier is transferred to output either in phase or  $180^\circ$  with respect to reference carrier oscillator.
- ❑ Balanced ring modulator circuit is shown in Fig.2.12.
- ❑ Digital voltage must be much greater than the peak carrier voltage for proper operation.
  - ❑ For logic 1: D1 and D2 are ON while D3 and D4 are OFF, carrier voltage across T2 is in phase with the carrier voltage across T1 or the reference oscillator.
  - ❑ For logic 0: D1 and D2 are OFF while D3 and D4 are ON, carrier voltage across T2 is  $180^\circ$  out of phase with reference oscillator.

# Transmitter of BPSK

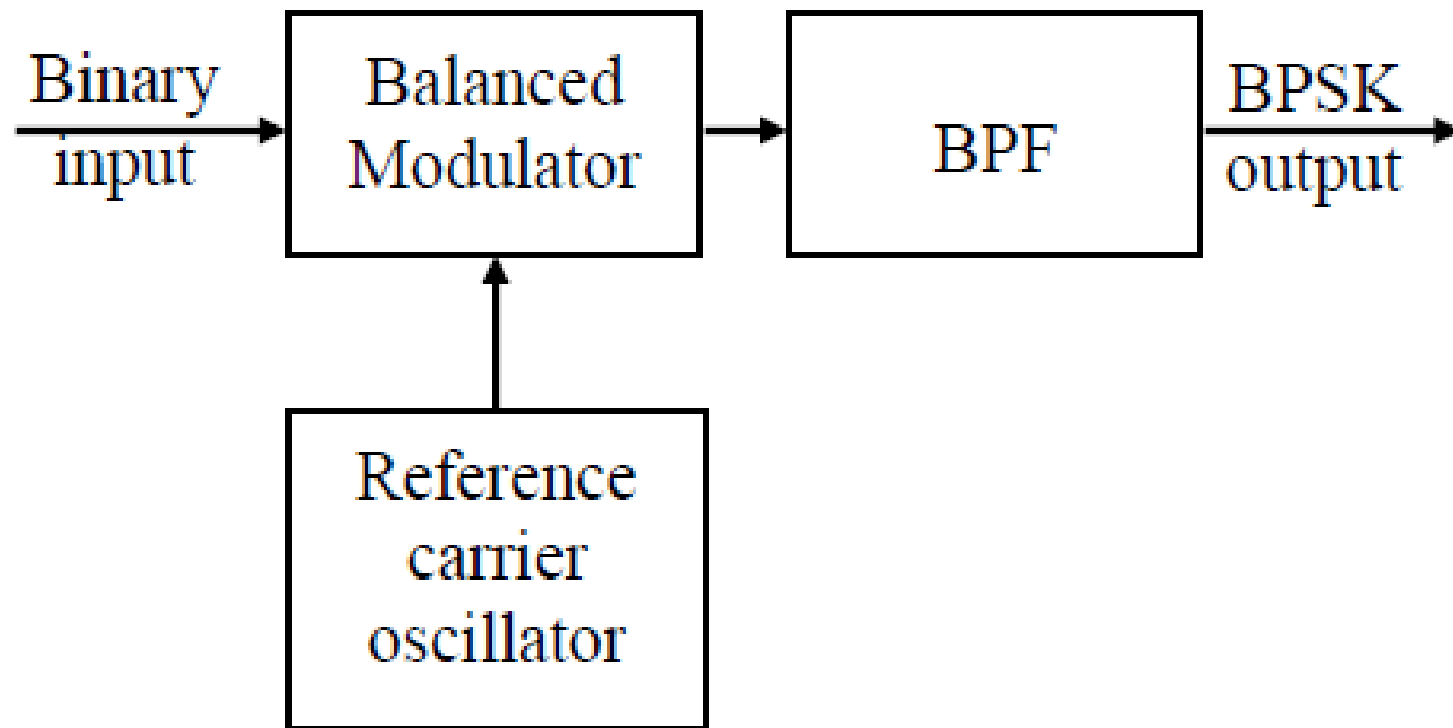


Fig.2.11 BPSK Modulator

# PSK Balanced Ring Modulator

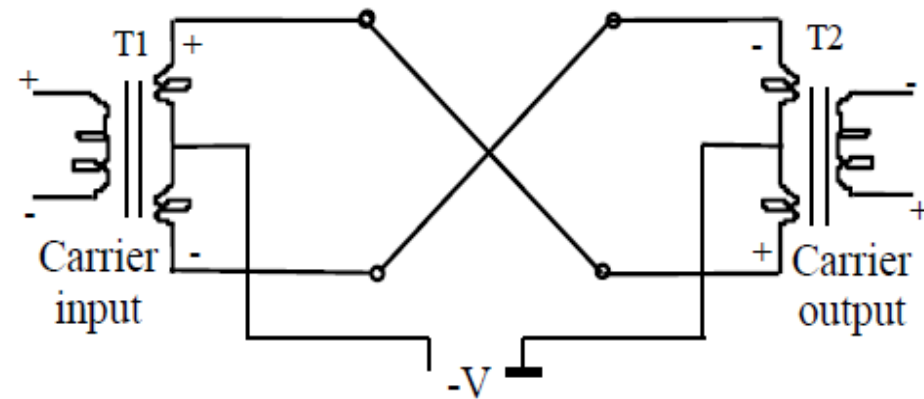
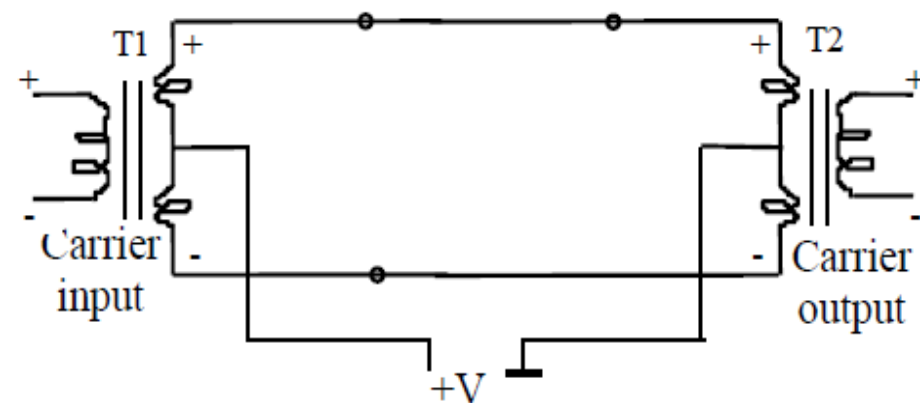
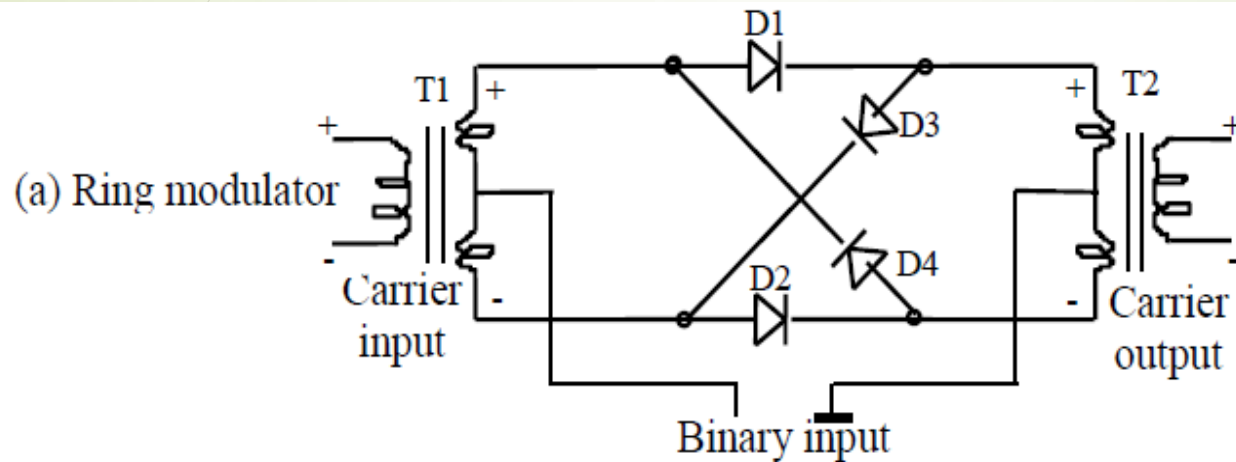


Fig.2.12 Balanced Ring Modulator

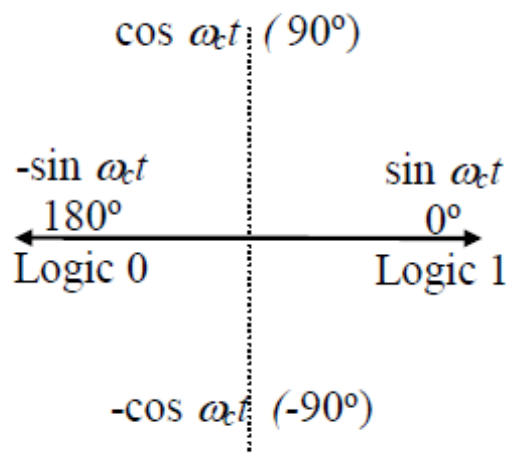
## Representation of BPSK

- Figure shows truth table, phasor diagram and constellation diagram for a BPSK.
- Constellation diagram is similar to phasor except that the entire phasor is not drawn.
- Only the relative positions of the peaks of the phasors are shown.

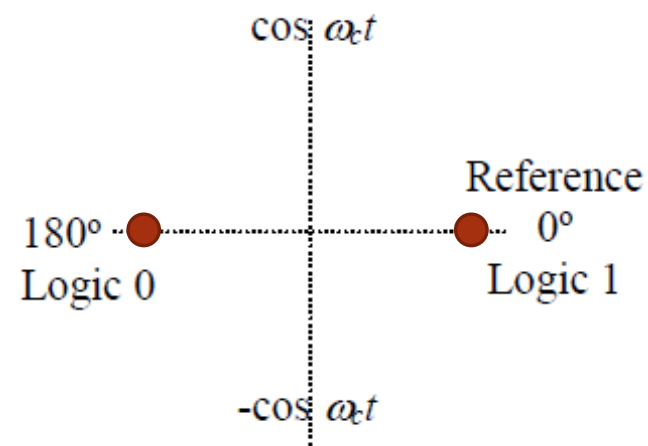
# Truth, Phasor, Constellation

Binary input	Output phase
Logic 0	$180^\circ$
Logic 1	$0^\circ$

(a) Truth table



(b) Phasor diagram



(c) Constellation diagram

Fig.2.13 BPSK Representation

# Band Width of PSK

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- ❑ Balanced modulator is product so carrier is multiplied by binary data (either +1 or -1).
- ❑ Also, widest bandwidth occurs when data is alternating 1/0 sequence.

- ❑ Product modulator output of the BPSK is:

$$output = \sin \omega_a t \sin \omega_c t = \frac{1}{2} \cos(\omega_c - \omega_a)t + \frac{1}{2} \cos(\omega_c + \omega_a)t$$

- ❑ Consequently, minimum double-sided Nyquist bandwidth is:

$$B_{BPSK} = (\omega_c + \omega_a) - (\omega_c - \omega_a) = 2\omega_a = 2(f_b / 2) = f_b$$

- ❑ Minimum bandwidth to pass worst-case BPSK equals input bit rate.

# PSK Reception

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- Simple block diagram of BPSK detection.
- Coherent carrier recovery circuit detects and regenerates carrier that is both frequency and phase coherent with the original transmit carrier.
- Balanced modulator output is the product of two inputs (BPSK signal and the recovered carrier).
- The LPF separates the recovered binary data from the complex demodulated spectrum.

# Detection of BPSK

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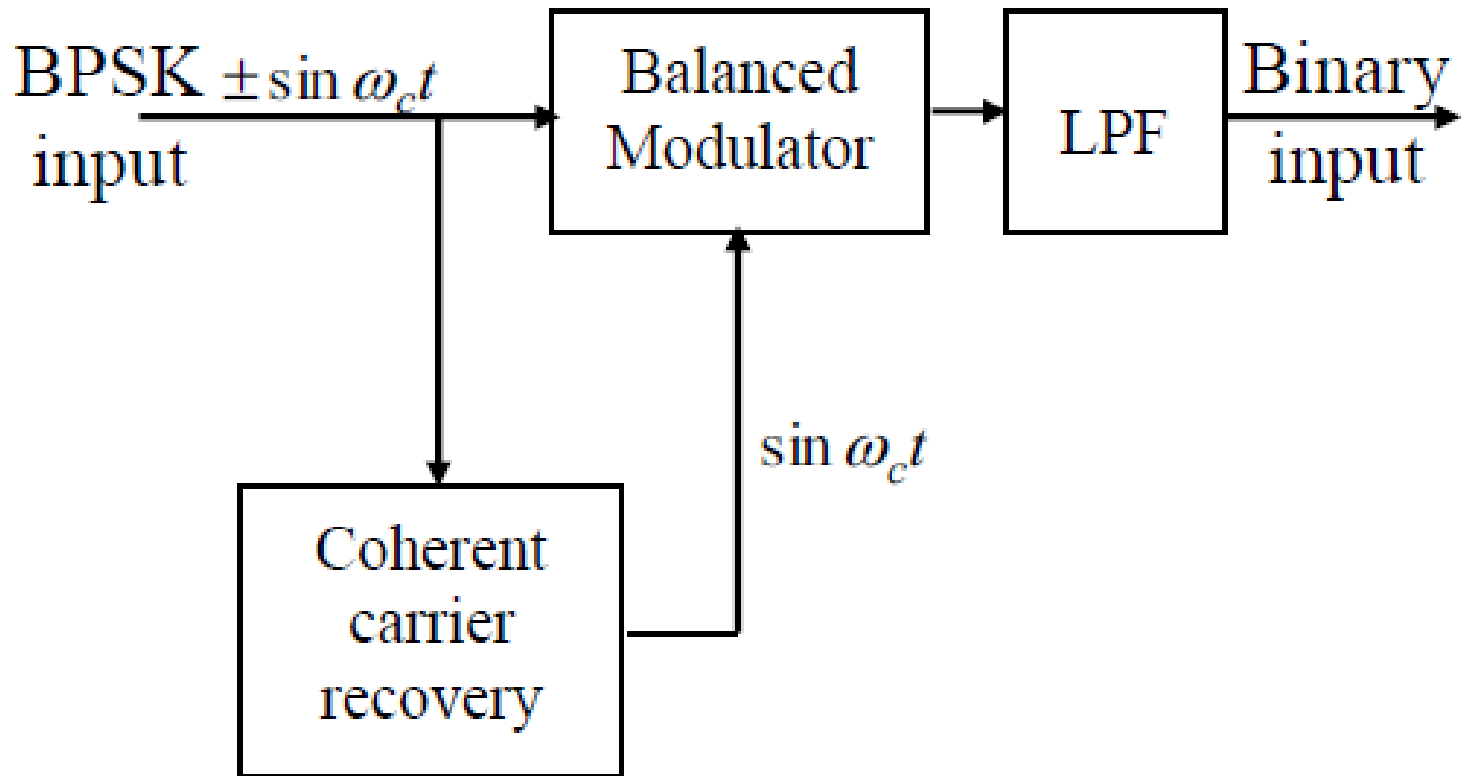


Fig.2.15 BPSK Receiver



# Demodulation Process

- For input  $+\sin \omega_c t$  (logic 1), balanced output is:

$$\text{Output} = \sin \omega_c t \sin \omega_c t = \sin^2 \omega_c t = \frac{1}{2}(1 - \cos 2\omega_c t) = \frac{1}{2} - \frac{1}{2} \cos 2\omega_c t$$

$$\text{Filter output} = +\frac{1}{2} \text{ dc voltage} \equiv \text{Logic 1}$$

- For input  $-\sin \omega_c t$  (logic 0), the output is:

$$\text{Output} = -\sin \omega_c t \sin \omega_c t = -\sin^2 \omega_c t = -\frac{1}{2}(1 - \cos 2\omega_c t) = -\frac{1}{2} + \frac{1}{2} \cos 2\omega_c t$$

$$\text{Filter output} = -\frac{1}{2} \text{ dc voltage} \equiv \text{Logic 0}$$

# **M-ary** **Phase Shift** **Keying**

# M-ary Encoding

- In  $M$ -ary, one of  $M$  possible signals may be transmitted during each signaling interval.
- It is advantageous to encode at a level higher than binary, e.g., 4PSK and 8PSK.
- Each possible transmitted signal of an  $M$ -ary message sequence is referred to as "symbol".
- Mathematically, the number of bits per symbol  $n$  is related to the number of possible signals  $M$  by:

$$M = 2^n$$

# Quadrature PSK

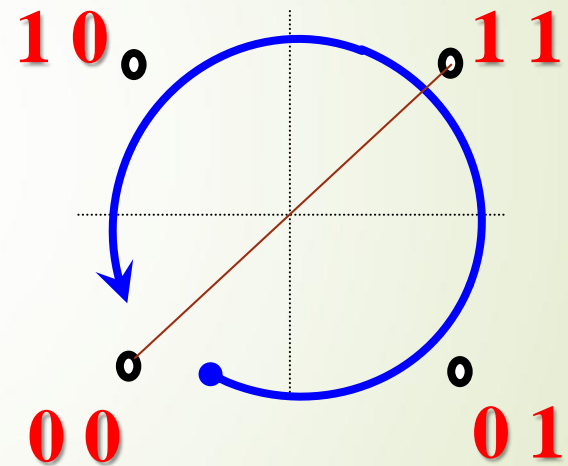
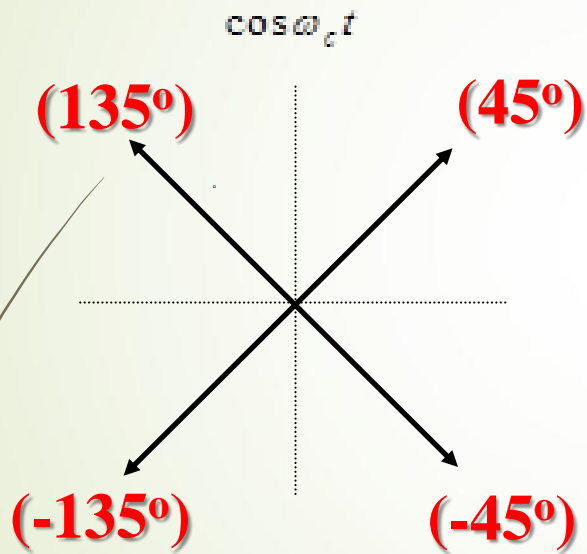
- QPSK, is another form of angle modulated, constant envelope digital modulation, and  $M = 4$  possible symbols.
- 4 phases are possible for a single carrier frequency.
- Binary input data are combined into groups of 2 bits called dibits.
- Each dibit code generates one of the four possible output phases.
- For each 2- bit, a single output change occurs. So, the output baud rate is one-half of the input bit rate.

# QPSK Truth Table

Inputs		Output
A	B	Phase
0	0	-135
0	1	-45
1	0	+135
1	1	+45

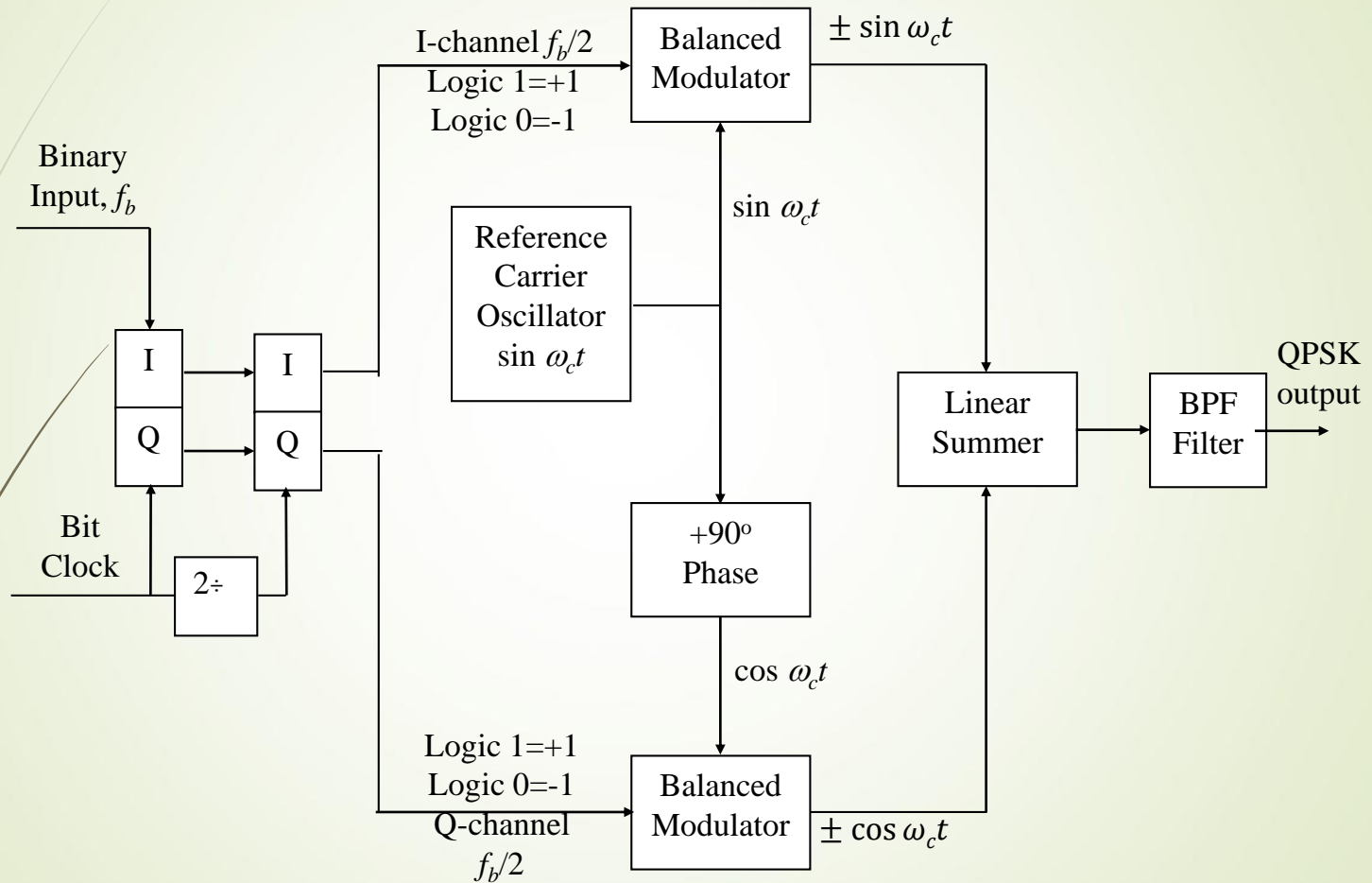
# QPSK

## Phasor Constellation



$$\begin{array}{r}
 00 \\
 \hline
 01 \\
 \hline
 11 \\
 10
 \end{array}$$

# QPSK Transmitter



# Transmitter Operation

- QPSK modulator is a two BPSK modulators combined in parallel.
- **Two bits are clocked into the bit splitter.**
- **After both bits have been serially inputted, they are simultaneously parallel outputted.**
- **One bit is directed to I-channel to modulate the carrier that is in phase with the reference.**
- **Other bit is directed to Q-channel to modulate carrier that is 90° out of phase or in quadrature with the reference.**
- **If linear summer combines the two quadrature signals, there are 4 possible phases as follows:**

$$\pm \sin \omega_c t \pm \cos \omega_c t$$



# Splitting to I and Q Channels

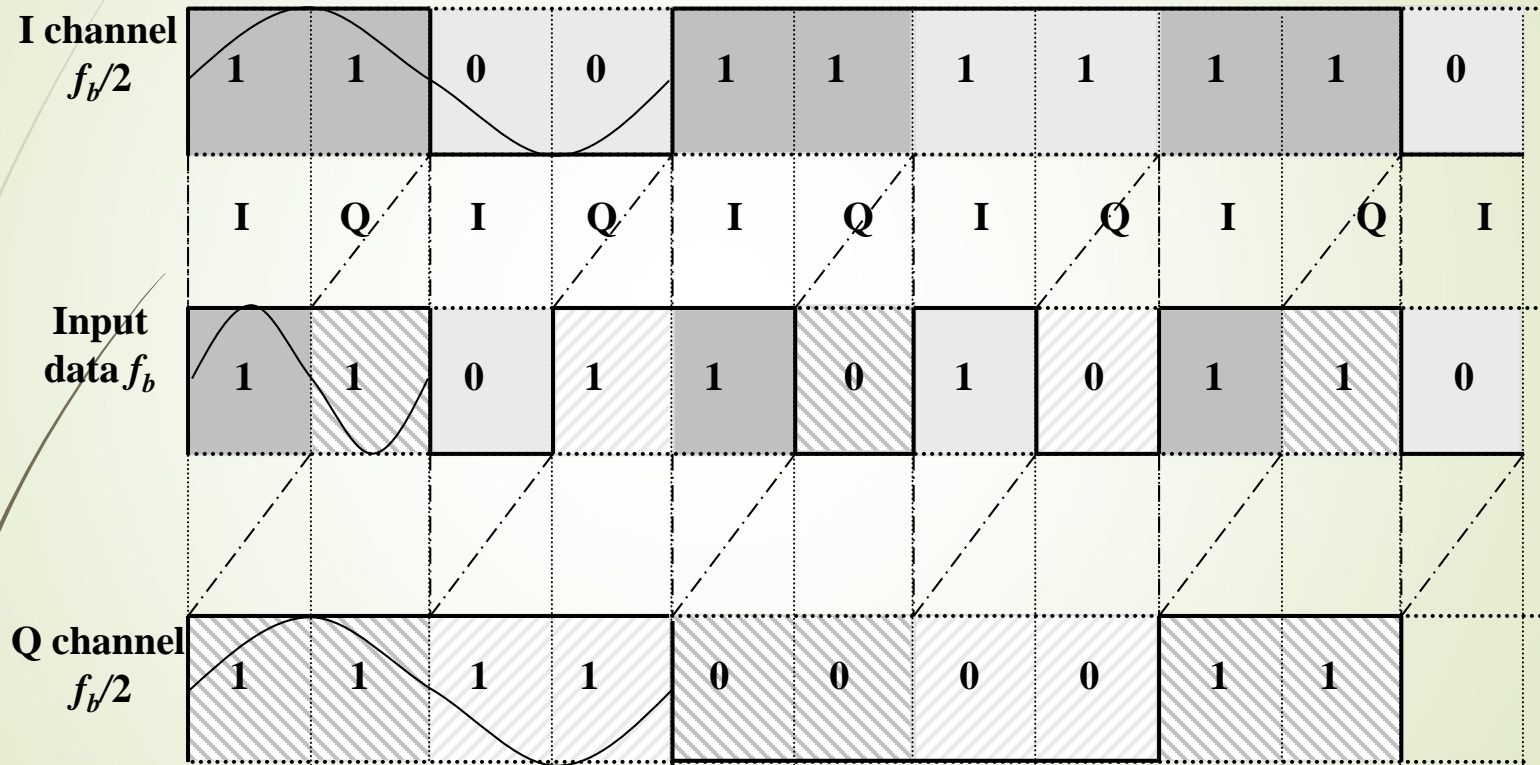


Fig.3.16: Highest Fundamental Frequency

# Bandwidth of QPSK

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- Input data rate  $f_b$  is divided into two channels.
- I or Q channel bit rate is  $\frac{1}{2}$  input rate, i.e.,  $f_b/2$ .
- Highest fundamental frequency at input of balanced modulators is one-fourth of input rate, i.e.,  $f_b/4$
- Balanced modulator product of I or Q channels:

$$\text{Output} = \sin \omega_a t \sin \omega_c t = \sin 2\pi \frac{f_b}{4} t \sin 2\pi f_c t$$

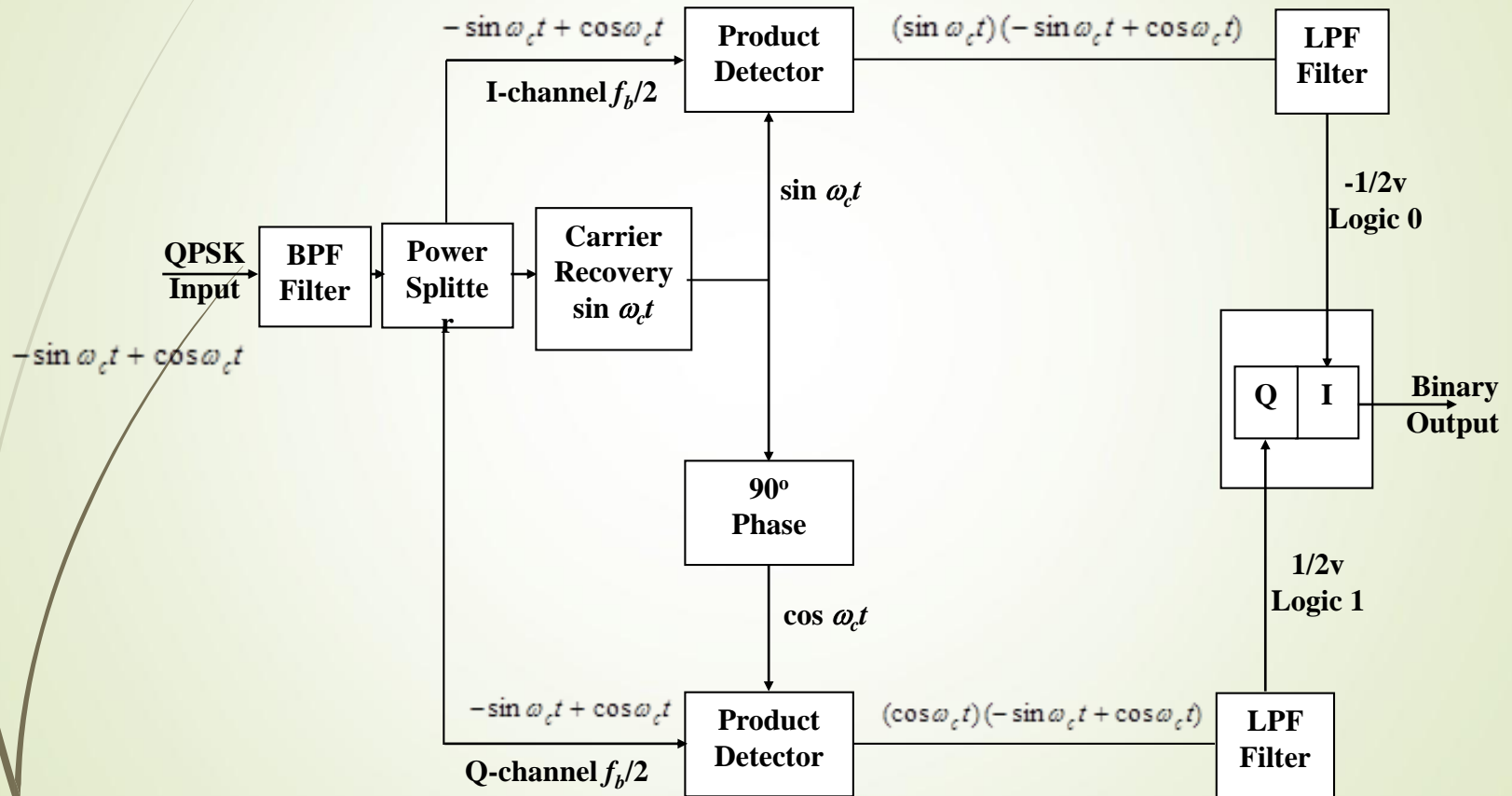
$$\text{Output} = \sin 2\pi \left( f_c - \frac{f_b}{4} \right) t \sin 2\pi \left( f_c + \frac{f_b}{4} \right) t$$

- So, output extends from  $f_c - f_b/4$  up to  $f_c + f_b/4$ :

$$BW_{QPSK} = f_c + \frac{f_b}{4} - \left( f_c - \frac{f_b}{4} \right) = \frac{f_b}{2}$$

- Minimum bandwidth of QPSK is less than incoming rate so that bandwidth is compressed to  $f_b/2$  only.

# QPSK Receiver



# Receiver Operation

- Power splitter directs QPSK signal into I and Q channels and carrier recovery circuit.
- Carrier recovery circuit reproduces the original transmit reference carrier.
- QPSK signal is demodulated in I and Q channels through product detectors.
- Detectors outputs are fed to combining circuit, to convert from parallel I and Q channels to a single binary output.

# Offset QPSK [OQPSK]

A modified form of QPSK where the bit waveforms on I and Q channels are offset or shifted in phase by one-half of a bit time.

- It can be implemented by adding a delay.
- In QPSK, change from 00 to 11 or 01 to 10 causes  $180^\circ$  shift in output phase.
- Since changes in I channel of OQPSK occur at midpoints of Q bits, there is never more than a single bit change in the dibit code,
- So,  $90^\circ$  shift in phase improves performance.
- Disadvantage: changes in phase occur at twice the data rate so bandwidth is twice.

# OQPSK Transmitter

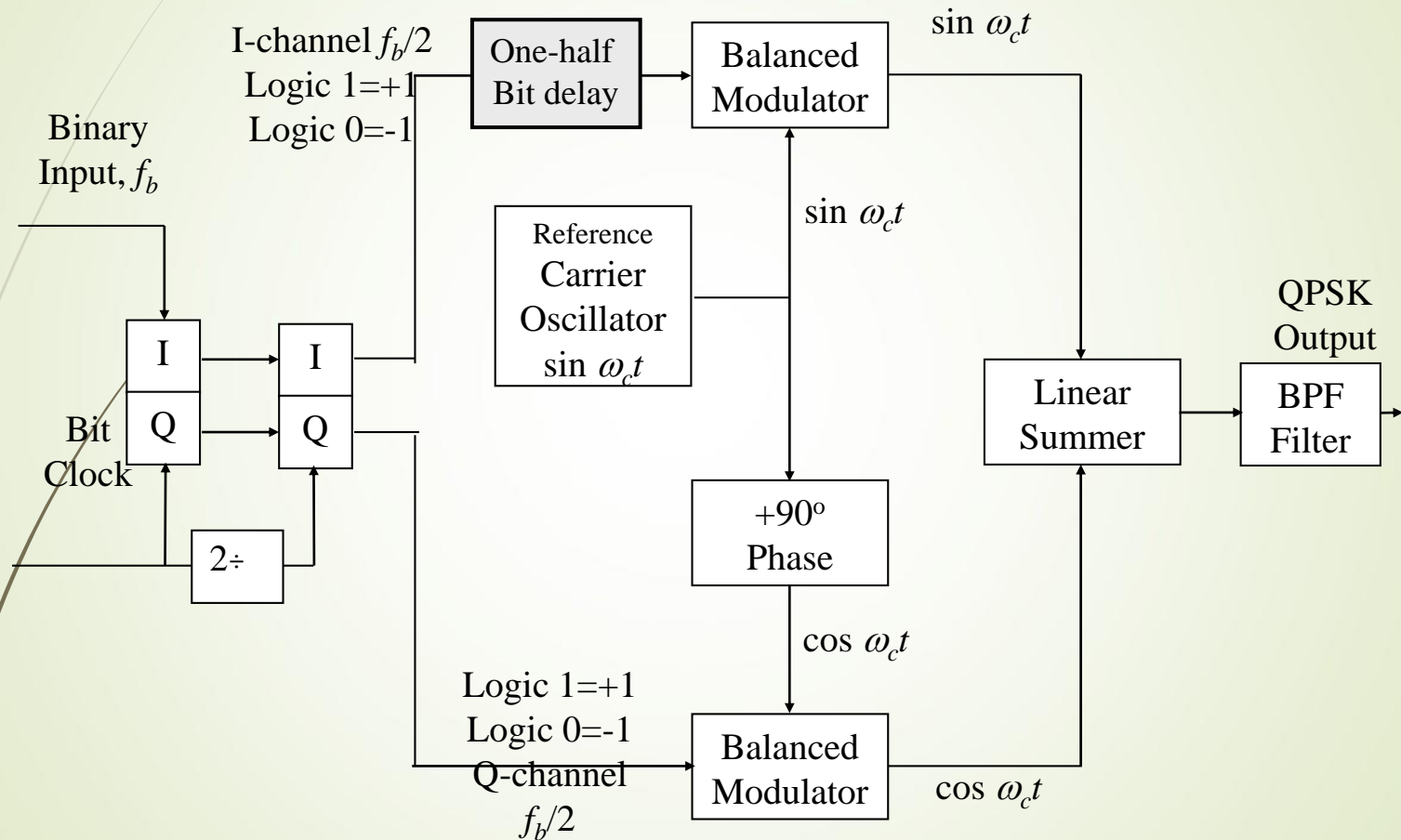
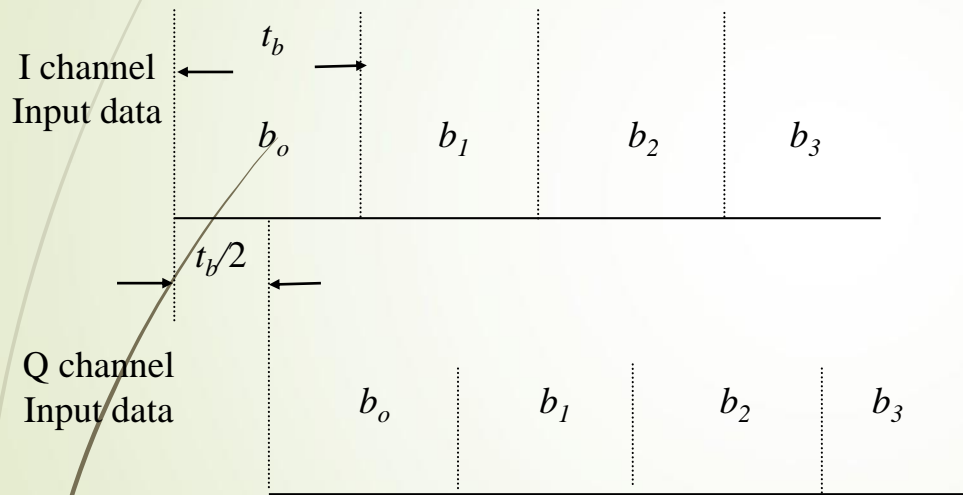
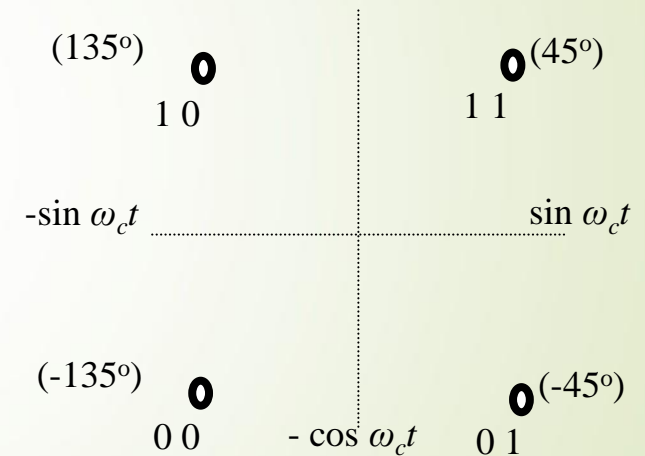


Fig.3.18: Offset QPSK Modulator

# Offset Delay Concepts



(a) Bit sequence alignment



(b) Constellation diagram

**Fig.3.19: OQPSK**

# 8 PSK



# Eight Phase PSK

## Phases of 8 PSK:

➤  $\Delta\phi = \frac{2\pi}{8} = \frac{\pi}{4} = 45^\circ$

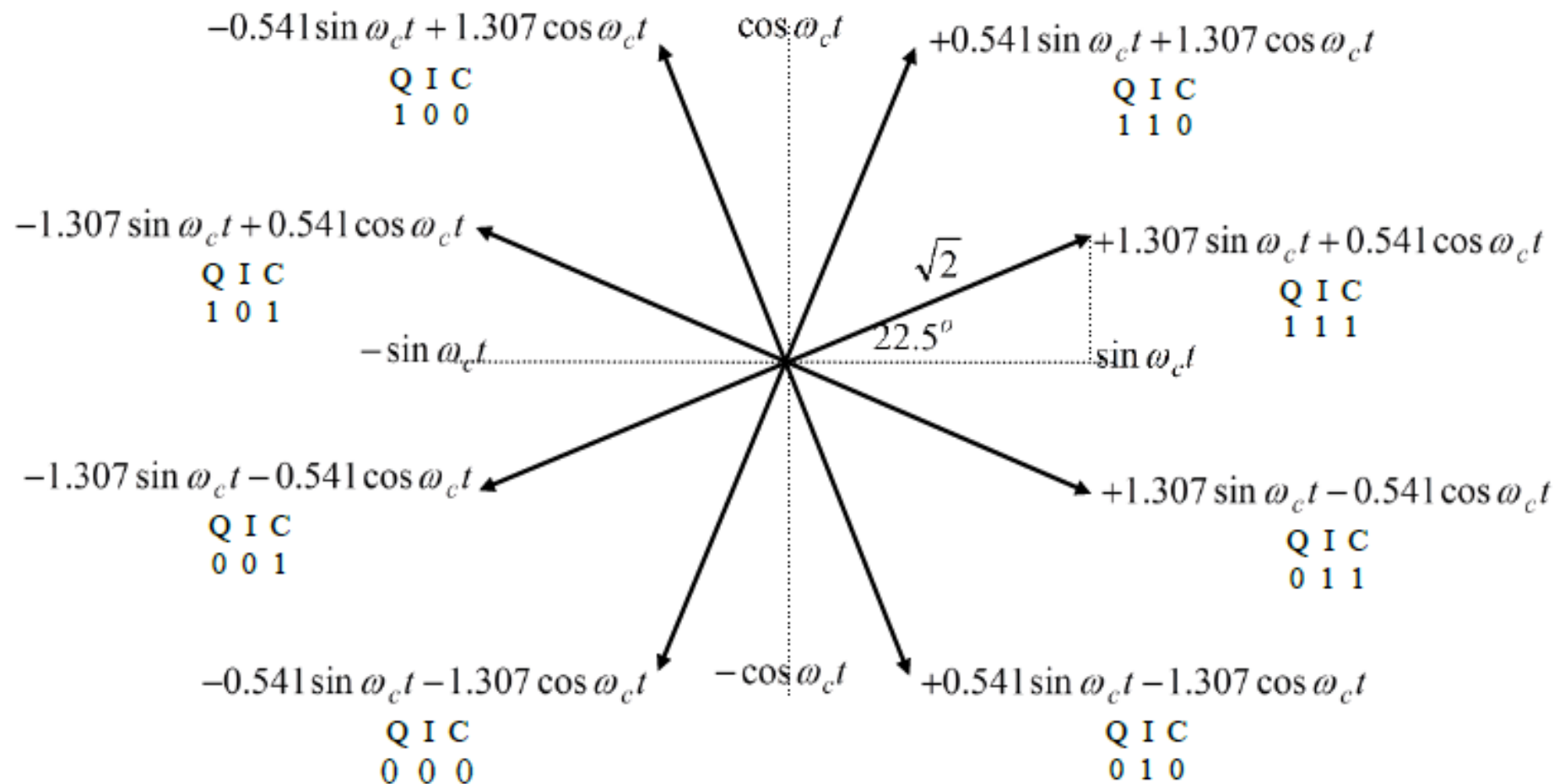
➤ First phasor =  $\frac{\Delta\phi}{2} = \frac{45}{2} = 22.5^\circ$

➤ Second = First +  $\Delta\phi = 22.5 + 45 = 67.5^\circ$

➤ Third = Second +  $\Delta\phi = 67.5 + 45 = 112.5^\circ$

➤ Last = First +  $(n - 1)\Delta\phi = 22.5 + 7 * 45 = 337.5$

# 8PSK Phasor

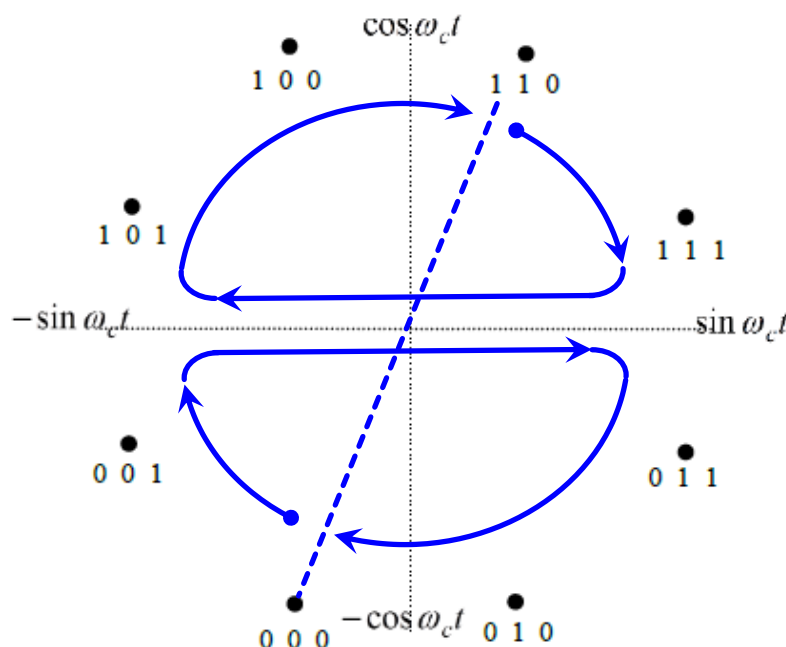


# 8PSK

## Truth - Constellation

Binary input			8-PSK Output phase
Q	I	C	
0	0	0	-112.5
0	0	1	-157.5
0	1	0	-067.5
0	1	1	-022.5
1	0	0	+112.5
1	0	1	+157.5
1	1	0	+067.5
1	1	1	+022.5

(b) The Truth Table of 8-PSK



(c) Constellation diagram of 8-PSK

0	0	0
0	0	1
<hr/>		
0	1	1
0	1	0
<hr/>		
1	1	0
1	1	1
<hr/>		
1	0	1
1	0	0

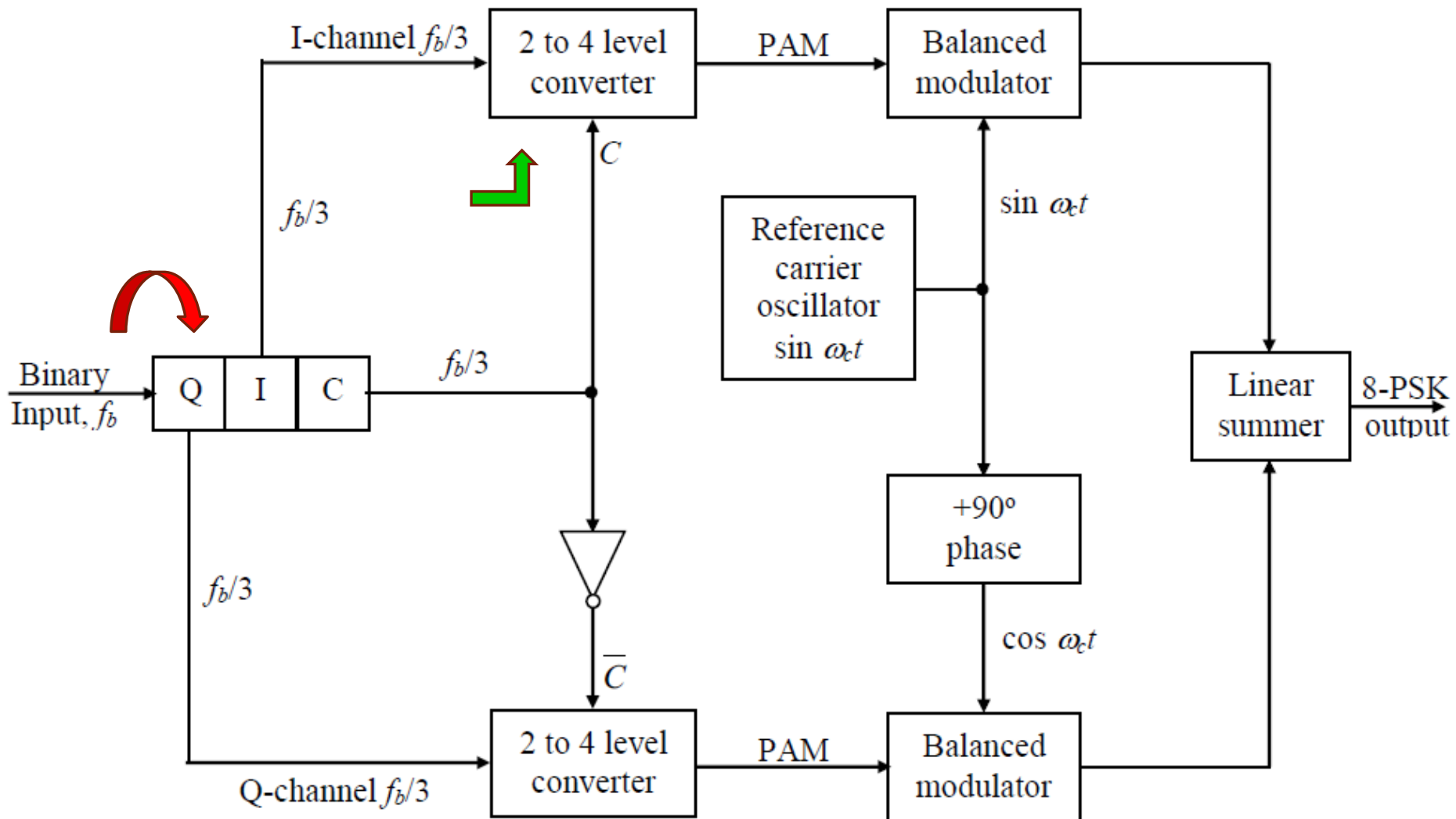
# 8 PSK Transmitter

## ➡ Serial to Parallel (Bit Splitter):

Converts binary input into 3 parallel channels:

- ➡ In Phase Channel, 'I'
- ➡ Quadrature Channel 'Q'
- ➡ Control Channel 'C'

# 8PSK Transmitter



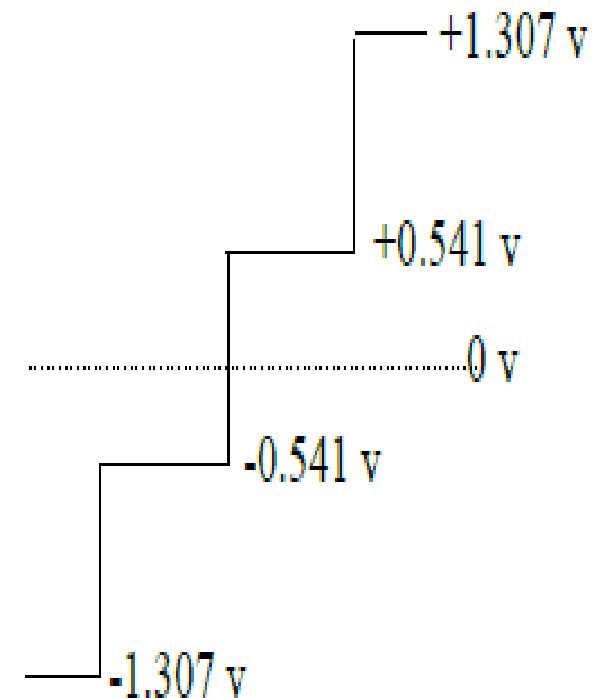
# 2-4 Converter

- Digital to Analogue Converter
  - I and C bits are inputs of DAC of I-Channel
  - Q and  $\bar{C}$  are inputs of DAC of Q-Channel
- With 2 parallel input bits, 4 output voltages are possible
  - I or Q bits gives polarity of analogue output
  - C or  $\bar{C}$  bits gives magnitude of output:
    - Logic 1: Magnitude is 1.307 v
    - Logic 0: Magnitude is 0.541 v

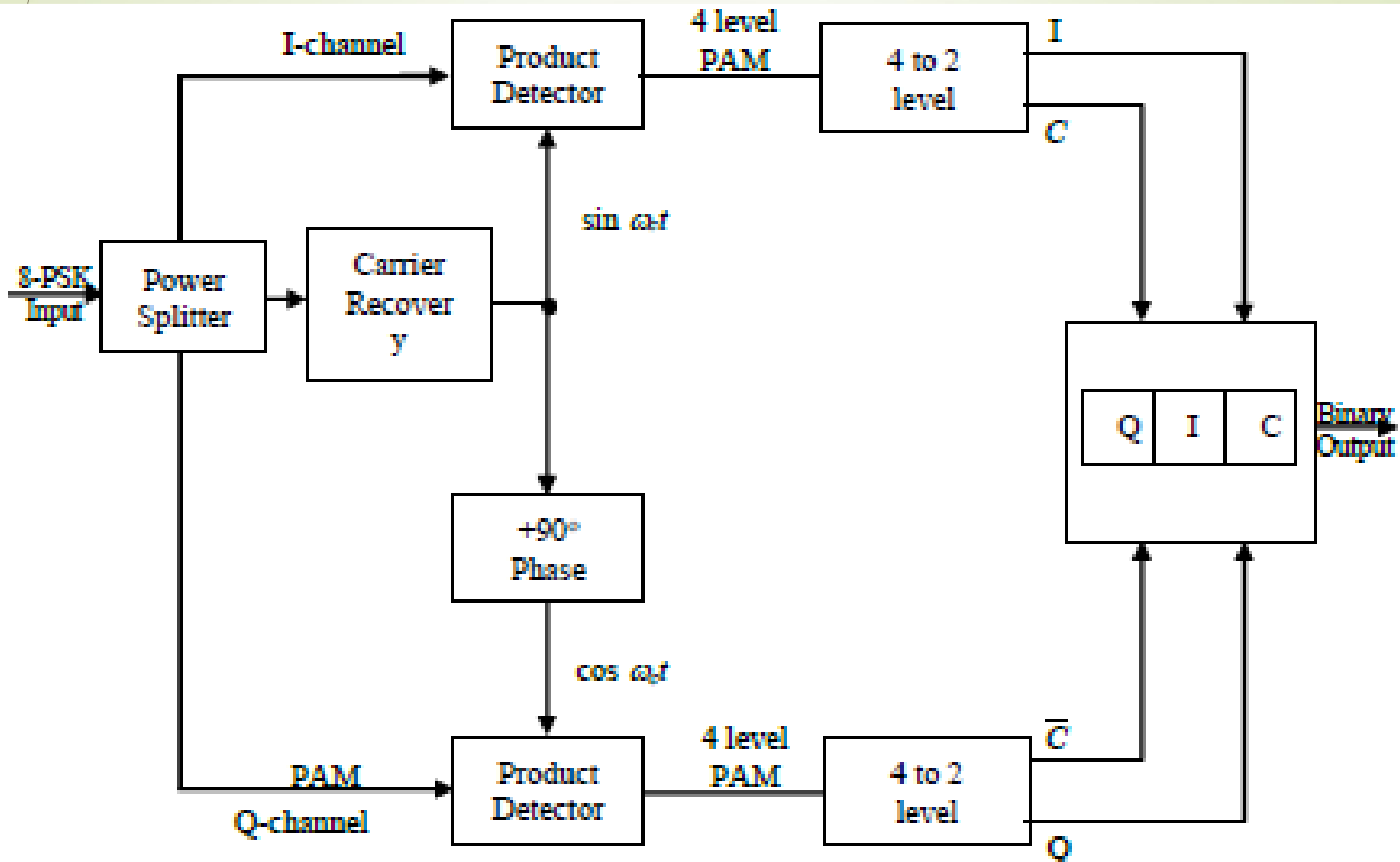
# 2-4 Levels Converter

I	C	Output
0	0	-0.541v
0	1	-1.307v
1	0	+0.541v
1	1	+1.307v

Q	$\bar{C}$	Output
0	1	-1.307v
0	0	-0.541v
1	1	+1.307v
1	0	+0.541v



# 8-PSK Receiver





# Receiver of 8 PSK

65

- Power splitter directs the 8-PSK signal to the I and Q product detectors and carrier recovery circuit.
- Carrier recovery circuit reproduces the original reference oscillator signal.
- Incoming 8-PSK signal is multiplied with recovered carrier in I channel and quadrature carrier (after  $90^\circ$  phase shift) in Q channel.
- Outputs of product detectors (4 level PAM signal) are fed to 4-to-2 level analog-to-digital converters.
- Outputs from I channel 4-to-2 level converter are I and C bits while those from Q channel are Q and C.
- Parallel-to-serial logic circuit converts I/C and Q/C bit pairs to serial I, Q, and C output data stream.

# BW of 8 PSK

- Data are divided into three channels, bit rate in I, Q, or C is equal to 1/3 of input (i.e.,  $f_b/3$ ).
  - Highest fundamental frequency in I, Q, or C channel is equal to 1/6 of input bit (i.e.,  $f_b/6$ ).
  - Also highest frequency in either PAM is  $f_b/6$
- Balanced modulators output is the product of carrier and PAM signal:

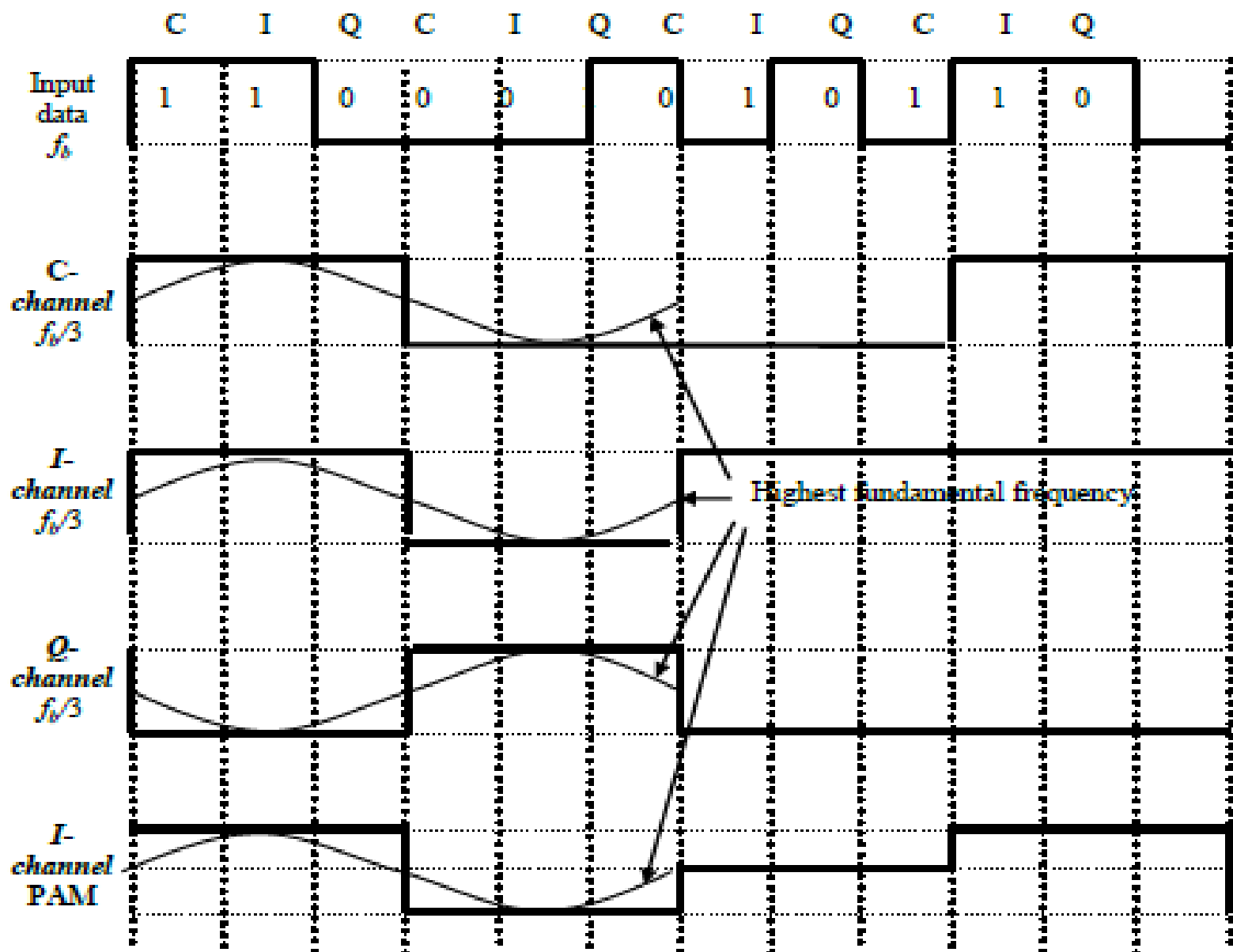
$$Out = A \sin 2\pi f_a t \sin 2\pi f_c t = A \sin 2\pi \frac{f_b}{6} t \sin 2\pi f_c t$$

$$Out == \frac{A}{2} \left[ \cos 2\pi \left( f_c - \frac{f_b}{6} \right) - \cos 2\pi \left( f_c + \frac{f_b}{6} \right) \right]$$

$$BW_{8PSK} = \left( f_c + \frac{f_b}{6} \right) - \left( f_c - \frac{f_b}{6} \right) = \frac{f_b}{3}$$

# Highest Fundamental Frequency

67



# 16 PSK

# 16 PSK

69

- There are 16 different output phases possible.
- Modulator acts on the incoming data in groups of 4 bits ( $2^4=16$ ), called quad bits.
- So, output phase does not change until 4 bits have been inputted into the modulator.
- Output baud rate and minimum bandwidth are equal to one-fourth of the incoming bit rate ( $fb/4$ ).
- Angular separation between adjacent output phases is only  $22.5^\circ$ .
- Therefore, the signal can undergo almost a  $11.25^\circ$  phase shift during transmission
- So, 16-PSK is highly susceptible to phase impairments in the transmission medium.

# **QAM**

# **Q**uadrature

# **A**mplitude

# **M**odulation

# QAM

- QAM is a form of digital modulation, the information is contained in both the amplitude and the phase of the transmitted carrier.
- **8**-QAM is  **$M$** -ary encoding technique where  **$M$**  = 8.
- **8**-QAM output is not a constant-amplitude signal such as **8**-PSK.

■

# 8-QAM Transmitter

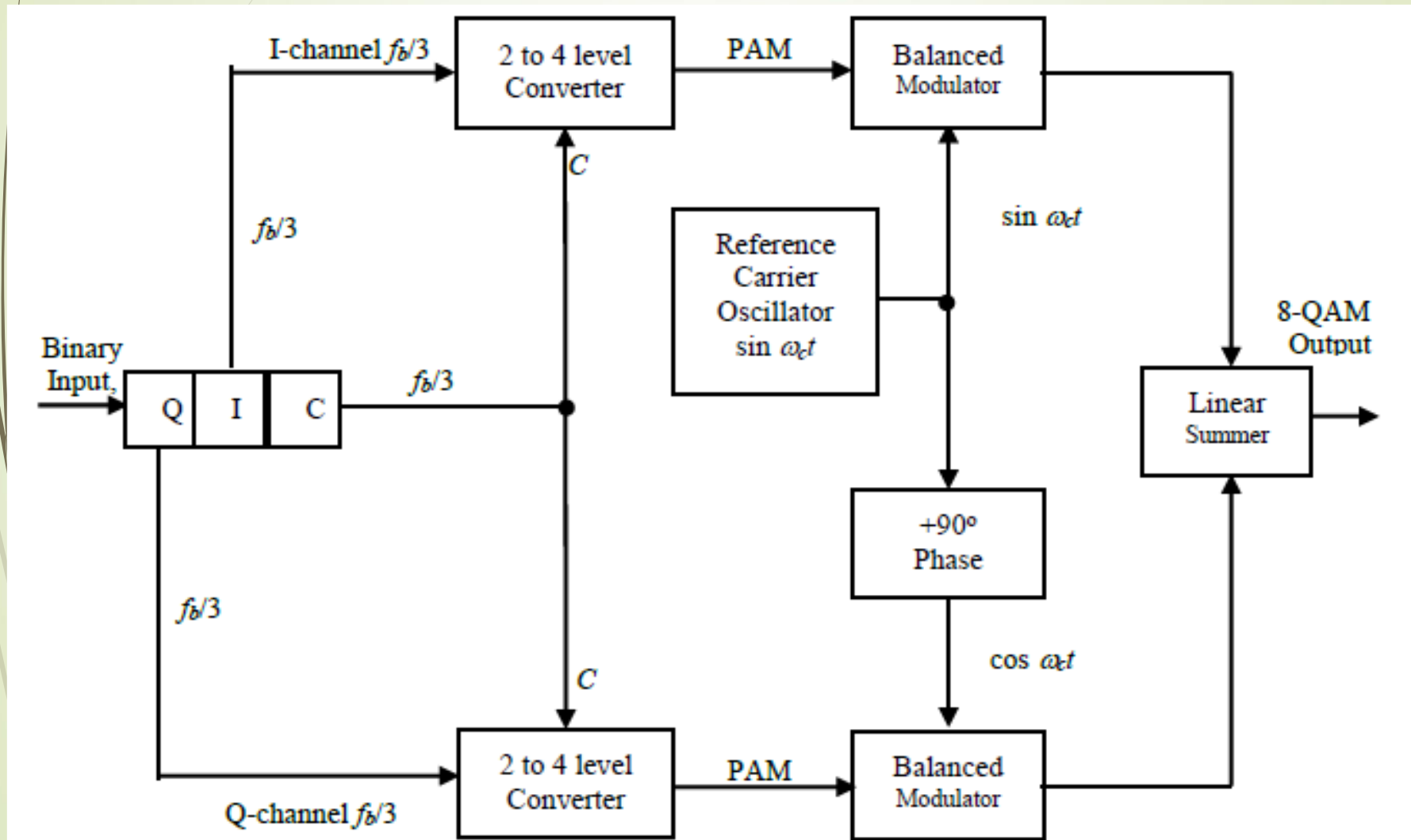
- 8-QAM differs from 8-PSK in the inverter between the C and Q.
- Data are divided into I, Q, and C channels; each with a rate  $f_b/3$ .
- I and Q bits determine the polarity of PAM signal at output of 2-to-4 level converters
- C channel determines the magnitude.
- 8-QAM output is not a constant-amplitude signal such as 8-PSK.



# 8-QAM Truth Table

BINARY INPUT			8-QAM OUTPUT	
Q	I	C	AMPLITUDE	PHASE
0	0	0	0.765 V	-135
0	0	1	1.848 V	-135
0	1	0	0.765 V	-45
0	1	1	1.848 V	-45
1	0	0	0.765 V	+135
1	0	1	1.848 V	+135
1	1	0	0.765 V	+45
1	1	1	1.848 V	+45

# 8-QAM Transmitter



# Comparison

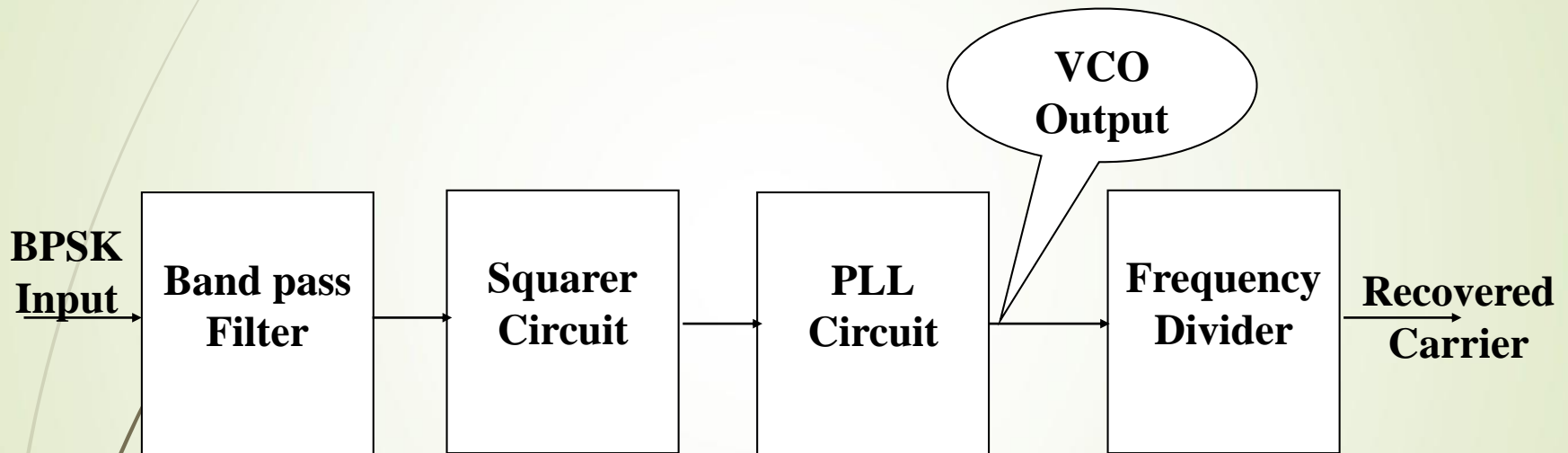
Table 2.2: Bandwidth Efficiency of Digital Modulation Techniques

Modulation	Encoding	Bandwidth, Hz	Baud rate	Efficiency, b/s/Hz
FSK	Single bit	$>f_b$	$f_b$	$<1$
BPSK	Single bit	$f_b$	$f_b$	1
QPSK	Di-bit	$f_b/2$	$f_b/2$	2
8-PSK	Tri-bit	$f_b/3$	$f_b/3$	3
8-QAM	Tri-bit	$f_b/3$	$f_b/3$	3
16-PSK	Quad-bit	$f_b/4$	$f_b/4$	4
16-QAM	Quad-bit	$f_b/4$	$f_b/4$	4

# Squaring Loop

- The received BPSK signal is filtered to reduce the spectral width of noise.
- Squaring circuit removes the modulation and generates the second harmonic of carrier.
- This harmonic is phase tracked by PLL.
- The frequency of PLL (VCO output) is then divided by 2 and used as a phase reference for the product modulators.

# BPSK Carrier Recovery



**Fig.2.27 Squaring Loop Carrier Recovery for BPSK**